

Gotharman's SPAZCboard6 For Little deFormer 3



Granular Analog Workstation

User Manual V01.43

Content Of This Manual

Introduction	9
In The Box	11
Getting Started	12
User Interface	19
Preset Select Screen	22
Operating LD3	26
Starting/stopping Sequencer/trigger parts manually	26
Selecting a Part	27
Note Tracks Steps view/edit	28
Selecting Note Track Steps	29
Setting Note Track Last Step	30
Mute/Unmute Note Tracks	31
Copy Morph Layer A to B	32
Selecting the next Preset	32
Selecting the previous Preset	33
Selecting Presets using step buttons	34
Preset Reload	35
Panic	35
Compare Preset	36
Shortcuts to Edit Pages	37
LD3 with Spazeboard6 Structure	39

Synth/Sampler Parts

[40](#)

Modulation Sources	41
Accessing The Synth Part Pages	48
Editing The Parameters	51
Parameters Randomizer	52
Trigger Setup Part 1-8	54
Trigger Setup Part 9-16	56
Zone Setup	58
Oscillator1/Sampler/Noise	59
Granular Modulators	71
Random Generators	73
Oscillator2/Ring Modulator	74
HPF and LPF – Analog Filters	77
Digital Filter (DGF)	84
VCA	89
Envelope 1 and 2	93
LFO 1 to 16	97
Effects Processors	100
Insert Effects 1 to 8	101
List of Insert Effects	102
Insert Effects Select Page	103
Mix Modulation and Backwards	104
Chorus	107
Distortion	109
Bit Crush	111
Pitch Shifter	113
Resonator	116
Stretcher	118
FM	120
Glitch Shifter 1, 2 and 3	122
Pitch Shaper	124
FAT	126
Filters	128
Filters 2	131
Compressor	133
Expander	135
PitchShaper2	137
Delays	140
Variator	142
Time Stretch	144

Sample Pitch	146
Output Effects 1 and 2	148
List of Output Effects	149
Output Effects Select Page	151
Mix/Pan Modulation and Backwards	152
Delays	154
Granulator SQ	157
Variator	160
Reverb	162
Granulator (unsynced)	164
Xfade Granulator	167
Abstract0	170
Time Stretch	173
Sample Pitch	175

Sequencer

[177](#)

Entering The Sequencer	184
Sequencer Main	187
Clear Sequence	189
Note subtrack	190
Gate Time subtrack	193
Velocity subtrack	195
Position subtrack	197
Sub Position subtrack	199
Templates and Note Scales	201
Templates for Position subtrack	203
Note Track Mod page	205
Clear Note Track	207
Double Note Track	208
Realtime Recording of Notes	209
Step Recording of Notes	211
Audio Track Recording	214
Controller tracks	224
Controller Track templates	227
Controller Tracks CC page	229
Clear Controller Track	231
Double Controller Track	232
Controller Tracks Realtime recording	233
Recording Parameter Tweaks	236
CV Inputs Recording	237
Pitch Bend and CC recording auto setup	238
Controller Tracks Step Recording	240

Synth and Sequencer Morphing

[241](#)

More... Parameters/Setup		<u>242</u>
Common Settings	245	
Morph Setup	247	
Common Settings 2	251	
CV Inputs	252	
Initialize Preset	255	
Parameter Snap Mode	256	
VCF Tune	258	
FLASH Memory Check	260	
Delete Sample Bank	261	
Delete All Presets and Songs	262	
C.P.	263	
Preset/Song Mode	264	
VCF Tjek	265	
Display Update Rate	266	
Save Preset		<u>267</u>
Song Mode		<u>271</u>
Accessing Song Mode	272	
Song Edit page	275	
Song Select page	277	
Song Realtime Recording	279	
Save Song	<u>283</u>	
Initialize Song	287	
Copy/Paste		<u>289</u>
Sample Record and Edit		<u>292</u>
Recording a Sampling	293	
Edit a Sampling	301	
Adjusting Start/End Points	303	
Sample Chops	304	
Wave Chop System	308	
Deleting Last Recorded Sampling	316	
Deleting Other Samplings	318	
Graphical Sample Editing	319	
Wave Builder		<u>332</u>

USB		<u>335</u>
	USB Sample Preview	337
	Open A Directory	338
	Importing Files	339
	Importing Multiple Files	341
	Reload Multiple Files	342
	Import Samplings As Chops	344
	Make a New Directory	347
	Delete File From USB Drive	348
	Export Samples, Presets and Songs	349
	Updating LD3 Firmware	<u>351</u>
MIDI Specs		<u>358</u>
	Parameter CC Control	360
	List Of Parameter CC Numbers	362
	Effects Parameters 1, 2 and 3	365
	Controlling Parameters From Controller Tracks	366
Installing Spazeboard6		<u>367</u>

I hope that you will deform some great tracks

Introduction

Thank you very much for purchasing/consider to purchase a Gotharman's Little deformer 3 with SPAZEboard6.

The Spazeboard6 version of Little deformer 3 (sometimes shortened to LD3) are a multi-timbral granular workstation with 8 monophonic internal parts, and 8 polyphonic external parts, for controlling MIDI gear. The granular functions in it includes oscillator/sampler granular modulators and special granular effects.

Each of the 8 internal parts has a stereo oscillator/sampler, a digital filter with 16 different filter types, a VCA, 3 envelopes, 4 random generators and 2 granular modulators, a random one and a sequenced one. Part 1 to 6 has an additional set of HPF + LPF analog filters, and 2 insert effects. These parts each forms an enhanced PolySpaze voice. Additional modulation sources includes 16 global LFO's, 32 sequencer controller tracks, touch screen keyboard Y position and MIDI keyboard, velocity, aftertouch, pitch bend and CC's.

The samplers can hold up to 193 minutes of samplings/ maximum 8.192 samplings. Samplings can be chopped by level peaks, wave zero points, or in equally sized slices. Up to 64 chop points are possible for each sampling. Samples are stored in FLASH memory, and played back directly from this, so: No loading times!

Each part can be sent two Output Effect Processors. Pan modulation can control the send level to each of the 2 effects.

The 16 parts can be sequenced from a built-in sequencer, that has 16 note tracks, with up to 64 steps each, and 32 controller tracks, with up to 128 steps each. Each note track has a position track, that makes it possible to alter the position of each step, making polyphonic step sequencing and various direction modes possible. A sub position track is also available for micro timing. Realtime, step time and xox style recording are possible. Knob movements and MIDI CC's can be recorded, both in realtime and step time on the controller tracks. Both the note tracks and the controller tracks can also control external MIDI devices, with up to 128 notes of polyphony.

Audio tracks can be recorded on the note tracks of part 1 to 8. These can be instantly chopped for true deforming manipulation.

The sound generating parts, the LFO's, the effects processors and the sequencer has 2 layers of parameter settings, layer A and layer B, that can be independently adjusted, and morphed between, using the Morph and Seq Morph knobs.

All the parameter settings can be stored in any of 1024 rewritable preset locations, and recalled at any time.

1024 song locations are available, for programmed playback of presets, and for tracks mute automation.

4 CV inputs and 4 separate audio outputs can be (optionally) added to Spazeboard6.

I hope that you will enjoy your LD3 with Spazeboard6 for a long time, and deform a lot of great tracks.

In The Box

In the LD3 box should be:

- LD3 itself with SPAZEboard6 installed.
- A power supply –Multi plug –Works in most countries.

If any of these items are missing, please get in touch with Gotharman's.

Getting Started

Connecting:



On the right end panel of your LD3, you will find the power switch, connection for power supply, stereo audio outputs, USB, and MIDI in and out.

You would probably like to connect the audio outputs to a mixer or an amplifier, or anything else that ends out in a speaker/a set of speakers. Since LD3 doesn't have built in speakers, it just needs to be connected to something, that can transfer its amazing sound to you. These should be connected, using ¼" mono jack cables.

The left audio output (marked "AUDIO OUT L(hp)") doubles as a stereo headphone connector. Please make sure that nothing is connected to the right audio output, when plugging headphones into this connector.

If the LD3 touch screen keyboard and step buttons seems a bit too limited, you might want to connect a MIDI keyboard to MIDI in, in order to take full advantage of LD3's fully chromatically

playable sounds. It is also possible to connect anything that transmits a MIDI clock, if you would like the sequencer of LD3 to sync to the rest of your setup.

On MIDI out, MIDI clock, MIDI CC's from the LD3 edit knobs, and notes and CC's from its sequencer are transmitted. Connect any MIDI gear to this, that you would like to control from LD3.

To the USB connector, a USB drive can be connected.

This should be:

- Maximum 32 GB

- FAT formatted

With a USB drive connected, you can:

- Import, export and back up samples as .wav files

- Import deFormer .lds samples

- Import, export and back up LD3 presets and songs

- Update LD3

- PLEASE NOTICE: The included factory samples CANNOT be exported. So if you want to keep these, you should take care not to delete them. A USB stick with the factory samplings might be available in the future.

To import a .wav file from another device, it must be:

- Mono or stereo

- 44.1 KHz sample rate – LD3 will import other sample rates, but they will play back in a wrong speed

- 16 bit or 24 bit native PCM

- Standard wav's or broadcast wav's



On the left end panel of your LD3, you will find the stereo audio inputs. The optional extra separate audio outputs and the CV inputs are also found here, if installed.

Audio inputs L and R

The two rightmost 1/4" jack connectors, marked "AUDIO IN R L" are always the stereo audio inputs. On the Spazeboard6 version, these will always go through VCF 5 and 6.

Connect any line stereo/mono audio sources to the audio input, for sampling and/or processing through LD3's analog filters 5 and 6, and the effects processors.

AUDIO IN/OUT 3, 4, 5 and 6 are separate audio outputs from analog VCF 1, 2, 3 and 4.

CV inputs

If you start counting from the left, the 4 minijacks are CV input 1 to 4.

Connect any CV voltage source to the 4 CV inputs. Each input can be set up to match the voltage range of any CV source, up to +/- 15 volts. The CV inputs can be set up to modulate many

parameters and they can be set up to act as trigger sources. They also accept clocks, start/stop and reset pulses for the sequencer.

Starting Up

Connect the supplied power adaptor to the Power input, and to a 100V to 240V power source – Usually a wall socket.



It's a 9V, minimum 2.0A type with a 2.1 mm DC plug, with positive middle. The power supply on the picture is only for reference. The actual one might look different.

Some LD3's might have been shipped out with a power adaptor, that has multiple tips. If you have received one of these, you should use the tip with the blue ring, and make sure that the 2 parts are aligned to the text "Tip":

Please look at the picture, on the next page....



Turn it on

Push the “I” on the power switch. Your LD3 should now turn on.

The User Interface



LD3 has a highly sensitive and responsive capacitive touch display, 16 step/trigger/part select/quick jump buttons, a Morph Set button, a Func/Mute button, a Steps/Part button and a Sequencer Start/Stop button. It has 8 Edit/Quick Edit Knobs for controlling and editing parameters and sending MIDI CC's, a (synth) Morph knob, a Sequencer Morph knob and a volume knob.

Pushing the **Trigger 1-16** buttons, with Func/Mute and Steps/Part unlit, will trigger the respective LD3 part. Each trigger button will send a settable note number (Settable in the Synth "Trig" section). When a trigger is triggered, the button will light up.

The **Start/Stop** button will start and stop Sequencer playback. When the sequencer is playing back, the Start/Stop button will light up.

The **Morph Set** button, will toggle the parameters on any Synth and Sequencer page, between 2 layers of parameters, A and B. The **Morph** knob (**MIDI CC#1**) will morph between the two layers of Synth parameters, and the **Seq Morph** knob (**MIDI CC#2**) will morph between the two layers of Sequencer parameters.

When the **Func/Mute** button is held down, it is possible to **mute/unmute** the 16 note tracks, by pushing any of the 16 step buttons. When it is lighting up, the 16 step buttons functions as **Quick Jump** buttons, that will take you to the edit pages, which names are showed right above each step button.

When the **Steps/Part** button is held down, the 16 step buttons functions as **Part Select** buttons. Pushing any of these, will select part 1 to 16.

When it is lighting up, the 16 step buttons will show the **steps** of the selected sequencer note track. Steps that are on will light up, steps that are off will not light.

The **Volume** knob always adjusts the audio output volume.

The **Edit 1-8 Knobs** to the left of the display, adjusts the parameters on each edit page. On the **Preset Select screen**, they acts as modulation sources, that controls any parameters that has knob1 to 8 set as modulator, and transmits MIDI CC's. Any Edit Knob, that has not been assigned as a modulator to any parameter, acts as a **Quick Edit Knob**.

The Quick Edit Knobs controls:

- Edit Knob 1: Analog HPF filter cutoff (**MIDI CC#4**).
- Edit Knob 2: Analog HPF filter resonance (**MIDI CC#5**).
- Edit Knob 3: Analog filters FM (**MIDI CC#8**).
- Edit Knob 4: VCA envelope release time (**MIDI CC#9**).
- Edit Knob 5: Analog LPF filter cutoff (**MIDI CC#10**).
- Edit Knob 6: Analog LPF filter resonance (**MIDI CC#11**).
- Edit Knob 7: Analog filters Boost (**MIDI CC#12**).
- Edit Knob 8: The level of the selected part (**MIDI CC#7**).

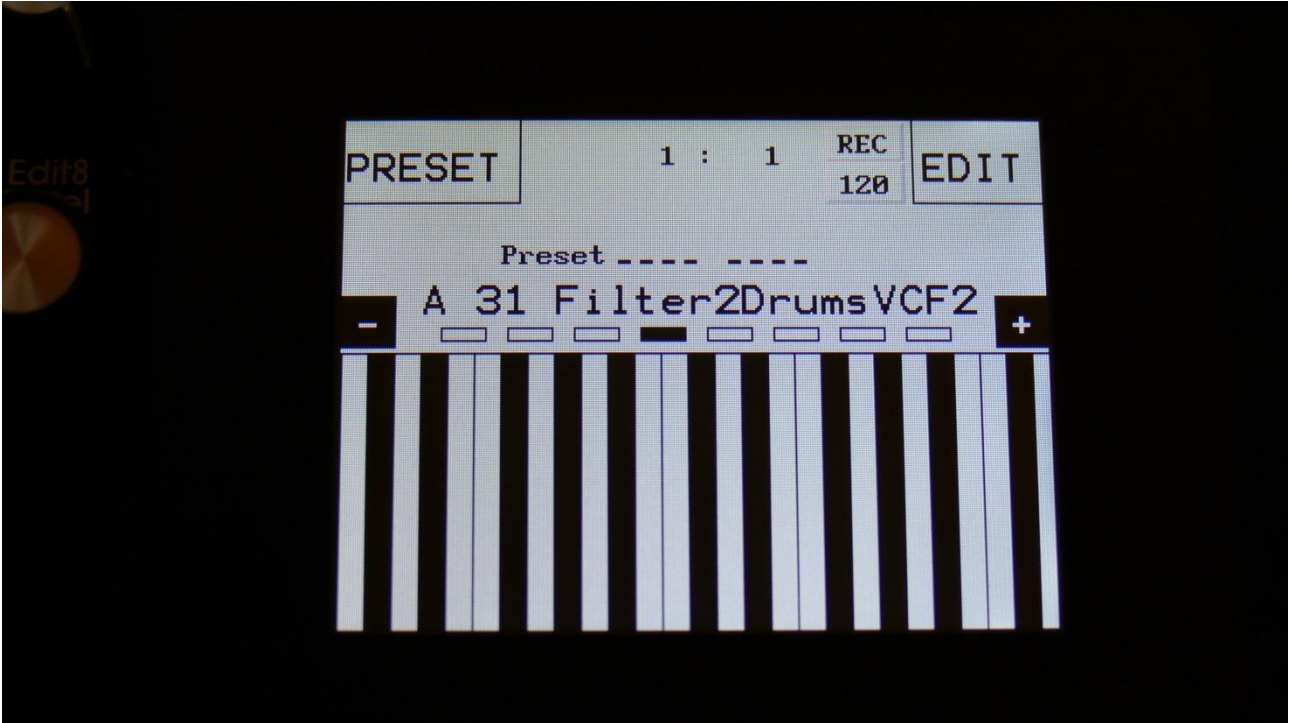
When the knobs is not assigned to any modulation destinations, they will work as Quick Edit knobs. When they work as Quick Edit knobs, and MIDI CC's are received, these will be received **per part**.

When the Edit Knobs are assigned as modulation sources, to any modulation destination, they will work **globally**.

The Morph knob and the Seq Morph knob will always work globally.

The Touch Screen Keyboard

The LD3 display is touch sensitive. The touch interface is used for navigating through the edit and settings pages, and in the bottom of most pages, a fully playable touch keyboard is present.



On the Preset and Song Select pages, it is, besides from playing notes on the touch keyboard, also possible to apply modulation to the sound, by placing your finger on different positions between the top and the bottom of the keyboard. This is referred to as Keyboard Y modulation. On any other pages, the keyboard only plays notes.

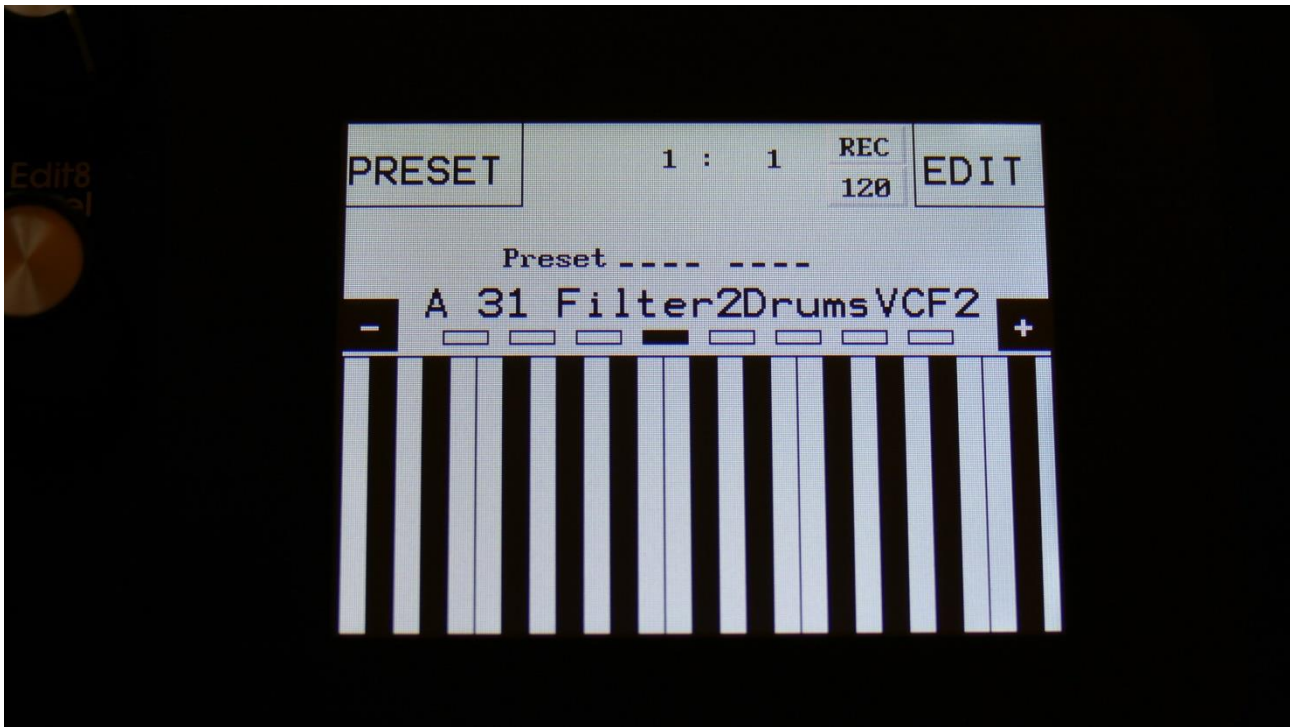
The touch keyboard is always controlling the selected part.

It is possible to select the keyboard octave, by touching any of the 8 squares just above the keyboard.

By touching the “-“ and “+“ buttons, just above the keyboard, it is possible to adjust the keyboard size. The size can be from 1 to 8 octaves.

The Y position modulation is transmitted, received and recorded as MIDI CC#16.

The Preset Select Screen



This is the first screen you will see, right after LD3's start-up screen, unless you left your LD3 in Song mode, the last time it was turned off. Here you can change preset, jump to LD3's edit and settings pages, and adjust the touch keyboard settings, as described on the previous pages.

On the top of this screen, the Sequencer bar/beat, that is currently being played back, is shown.

To the right of the bar/beat indicator, you will find a sequencer record indicator (REC) and a sequencer tempo indicator. When the REC indicator is red, the sequencer are in record mode. Touching the REC indicator, will switch the sequencer in and out of recording mode. Touching the tempo indicator, will make LD3 jump to the sequencer main page, where you can set the tempo. The tempo indicator will read "EXT", if LD3 is set to external MIDI sync.

Below the bar/beat indicator, it says "Preset", if LD3 is currently in preset mode, or "Song" if it is currently in song mode.

Below this, the number and name of the currently selected preset/song is shown.

Below the preset name/number, you will find the touch screen keyboard.

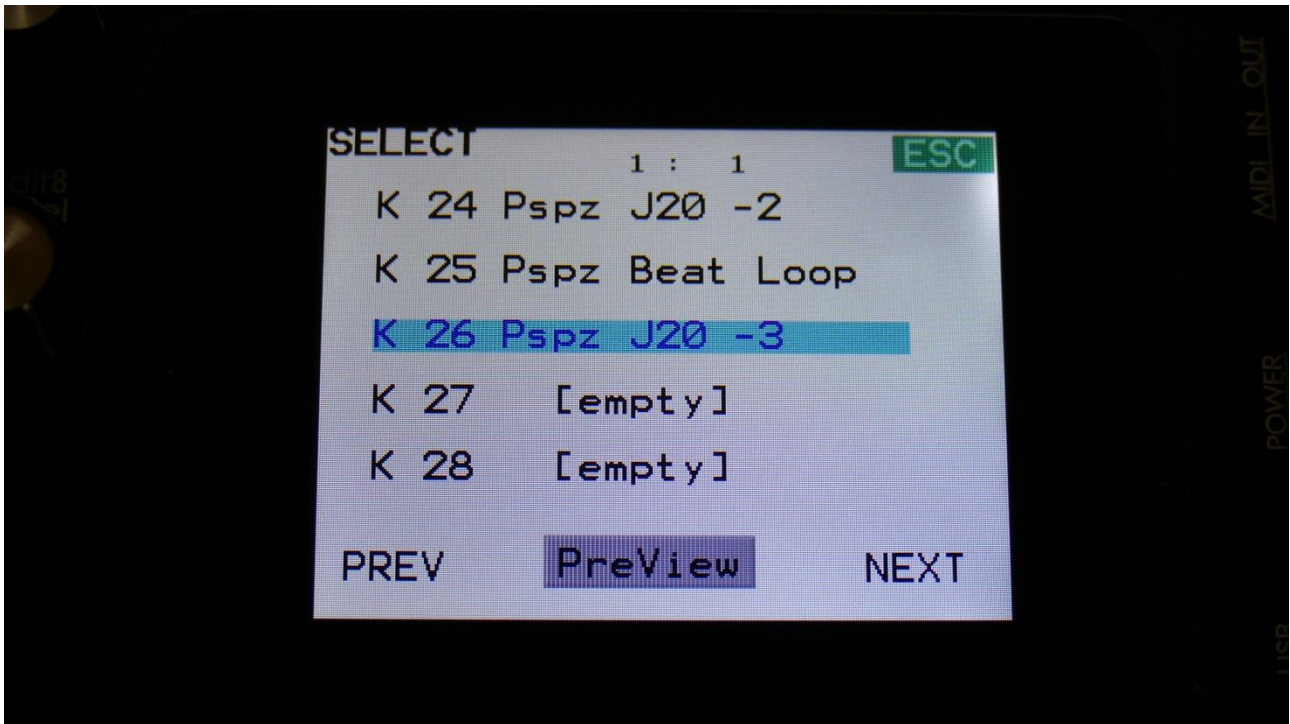
Right above the preset name, 8 small VU-meters are shown. These shows the activity of voice 1 to 8.

Touch the "EDIT" field in the upper right corner of the screen, to enter the edit and setup pages.

Touch the "PRESET" field, to select a memorized preset.

Selecting a preset:

Touch the "PRESET" field. A list of 5 presets near the currently selected preset, will now appear:



Touch "PREV" or "NEXT" to view the previous or next 5 presets, and finally touch the preset name of the preset you would like to select. 1024 presets can be selected, from A01 to P64.

LD3 will now jump back to the main Preset Select screen, and show the name of the newly selected preset.

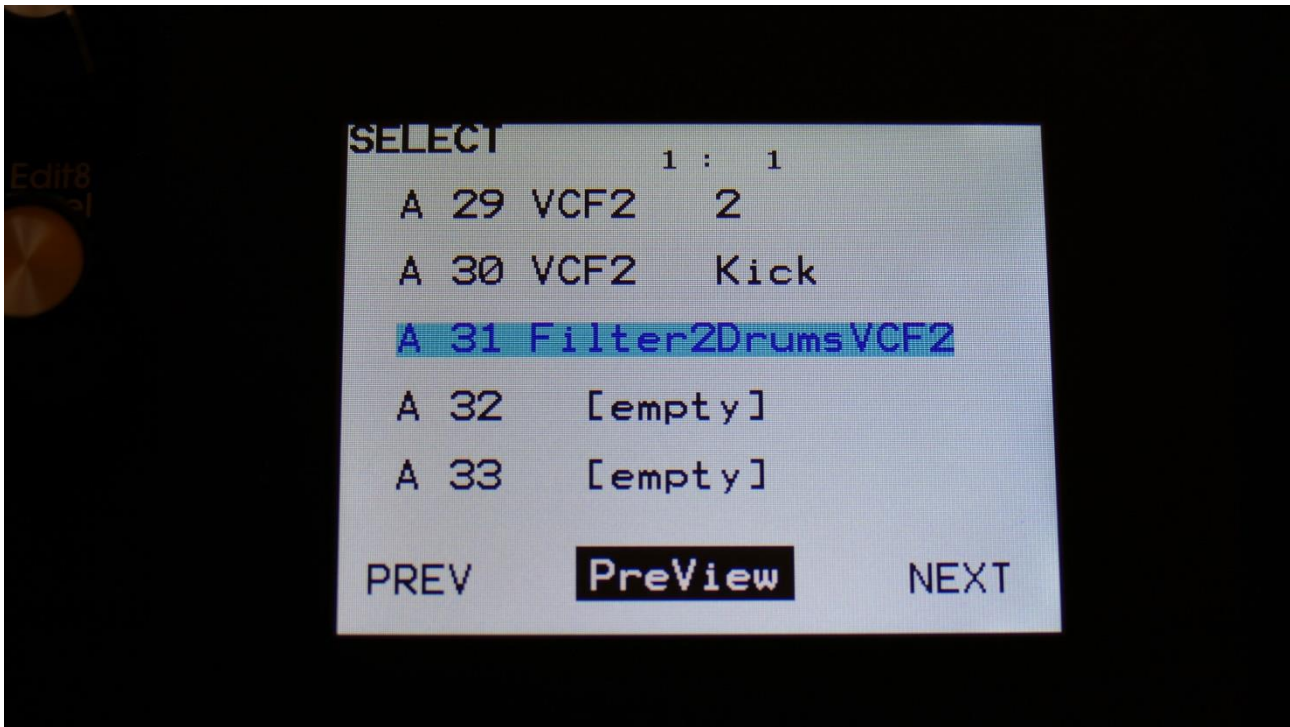
If the sequencer is playing back, the Start/Stop LED will now start to flash, and the text "NEXT:" will show right above the new presets name, awaiting track 1 to reach its start/end step. As soon as this happens, LD3 will switch to the newly selected preset, the Start/Stop LED will stop flashing, and "NEXT:" will disappear.

If the sequencer is not playing back, LD3 will immediately switch to the new preset, when you touch the preset name.

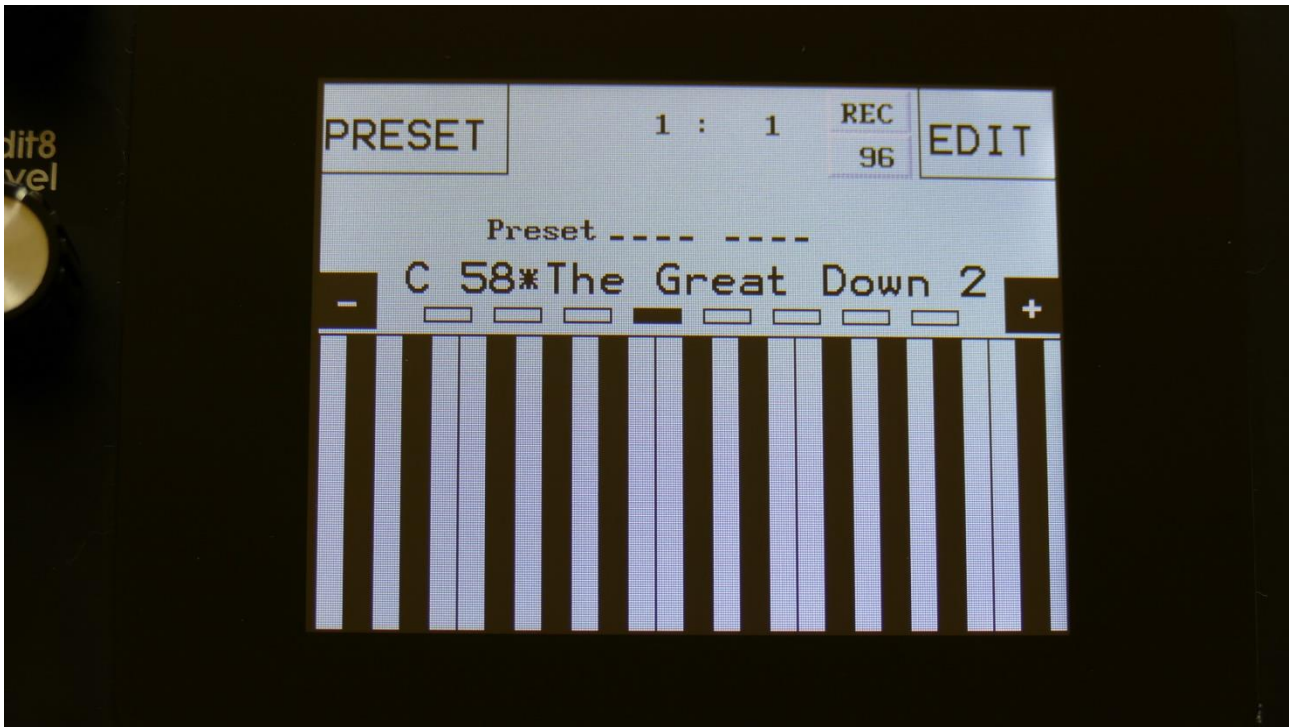
When LD3 is turned off, it will remember which preset was selected, and start up with this, when turned on again. It will also remember if it was in preset or song mode, and start up in the same mode, and if it was in song mode, it will also remember which song was selected.

Presets can also be selected, using the buttons. This is described in the next section “Operating LD3”.

If you activate the PreView mode, by touching the PreView field, so it turns black with white text, you can preview the presets, without LD3 jumping back to the main Preset Select screen.



If the sequencer is running, it will still wait for track 1 to reach step 1, until it jumps to the next preset. This is indicated by the Start/Stop LED flashing.

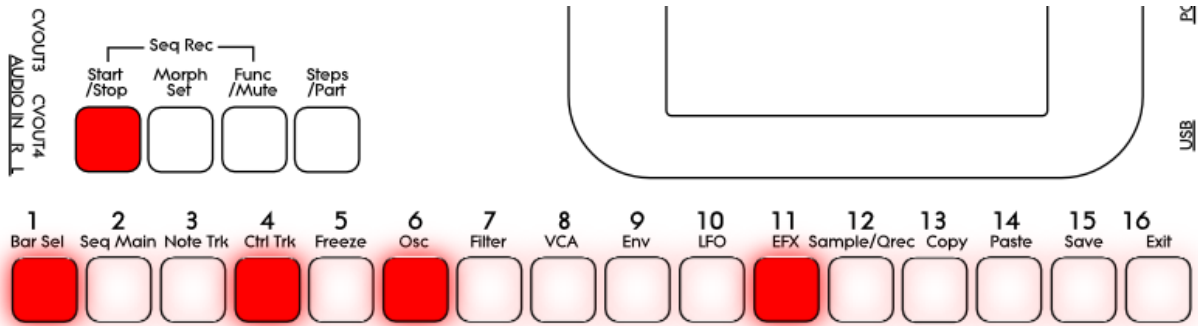


If any parameters of the preset has been edited, and the edits has not been stored, a “*” will appear right between the preset number and the preset name.

When this appears, you will need to save your preset, in order to keep your edits. Please see how to do this in the “Save Preset” section, later in this manual.

Operating LD3

Starting and stopping the Sequencer, and trigger parts manually:



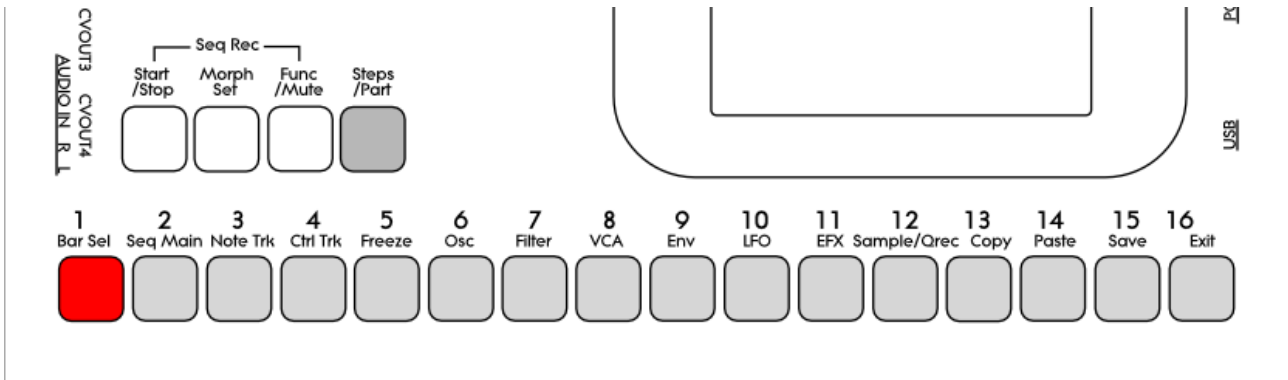
To start the sequencer playback, push and release the Start/Stop button, so that it lights up.

The sequencer will now start to play back. When the Func/Mute and the Steps/part buttons are not lighting up, every time a part is triggered to play back, the corresponding step button will light up shortly.

To stop the sequencer, simply hit the Start/Stop button again, so that it is no longer lighting up.

To trigger the 16 parts manually, make sure that neither the Func/Mute button or the Steps/Part button is lighting up, and then push any of the 16 step buttons, to trigger the sounds that are programmed on each part.

Selecting a part:



Push and hold the Steps/Part button.

The selected part number will now be shown, by one of the 16 step buttons lightning up. The number above the step button, is the part number that is currently selected.

To select another part, while still holding down the Steps/Part button, push any of the 16 step buttons.

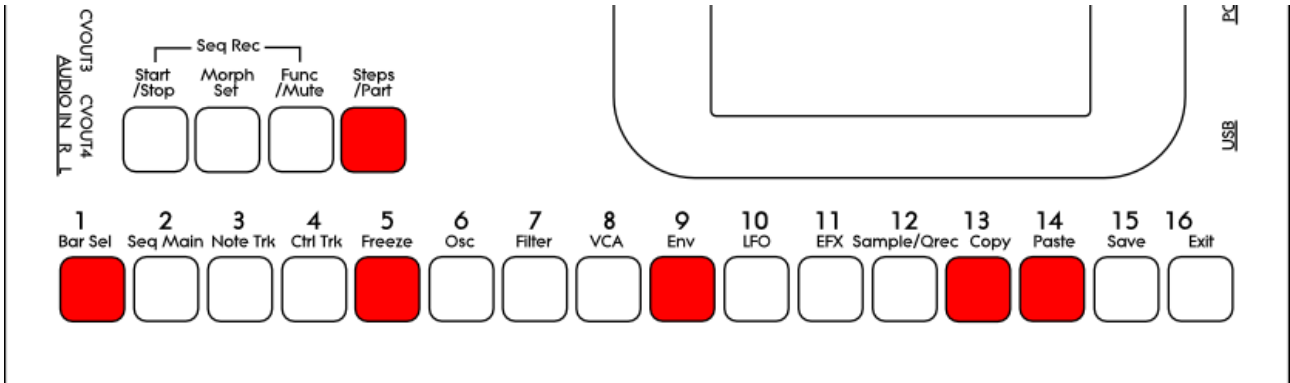
The selected part, is the part which parameters will be shown on the display, when entering the edit pages.

Selecting a part, also selects the equally numbered Sequencer Note Track.

When entering the Sequencer Controller tracks 1 to 16, track 1 to 16 is selected in the same way.

When entering the Sequencer Controller tracks 17 to 32, 1 will equal 17, 2 will equal 18, and so on.

Sequencer Note Track Steps view/edit:



First, select the part, for which you would like to view/edit the note steps, as described earlier in this manual. You can, of course, select another part at any time, also after you have entered note step edit mode.

To enter note step edit mode, push and release the Steps/Part button. This should now light up. Any note sequencer steps that are switched on to play back, will now also light up on the 16 step buttons.

If the sequencer is running, the light state of each step button will be reversed, when a step is playing back.

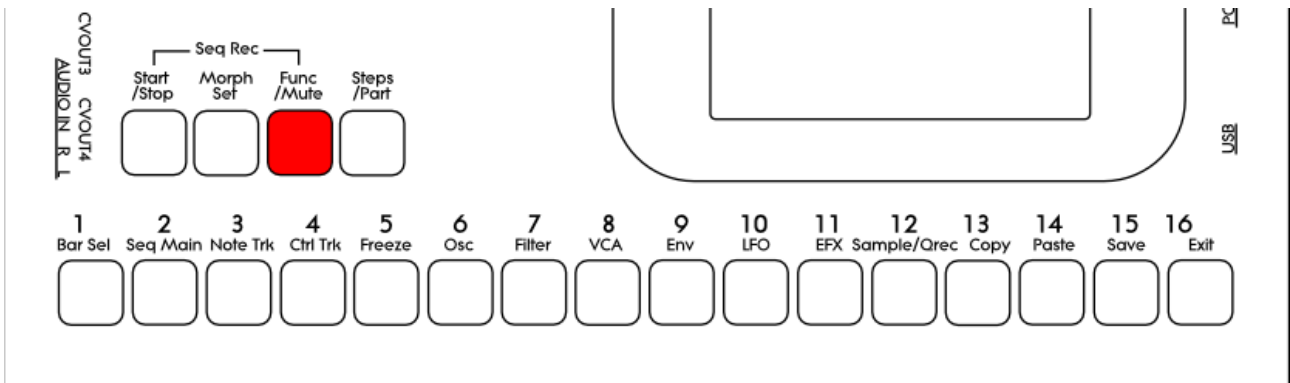
To switch a step on or off, simply hit the corresponding step button, as it will toggle its state.

When the sequencer are in recording mode (the Start/Stop button is flashing), pushing any of the step buttons, will set record mode to step mode, and select this step for recording. The step button will now flash.

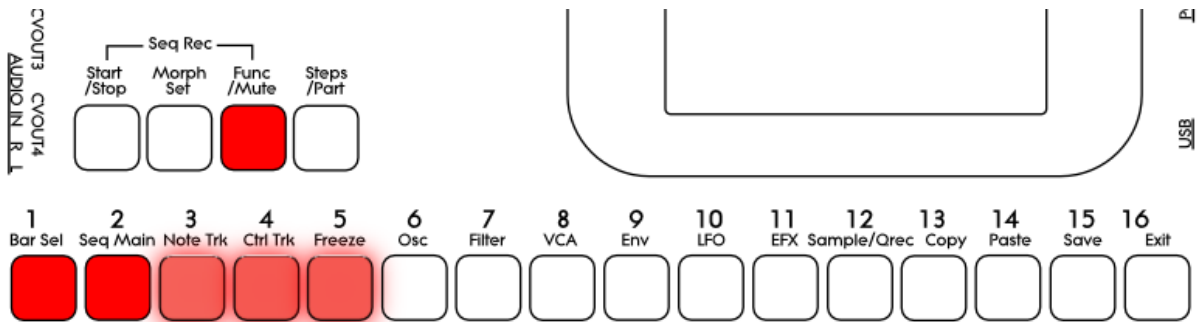
When using the LD3 sequencer as a usual step sequencer, the steps will play back from left to right. In this case, step 1 plays back at position 1, step 2 plays back at position 2 and so on. On the LD3 sequencer, it is though possible to break this pattern, and make each step play back on any position, using the position subtrack. It is even possible to make more steps playing back at the same position, for polyphonic step sequencing. But more on that in the final manual...

Only 16 steps are shown at a time. The note tracks of LD3 has 64 steps. Please read on, to experience how to switch between step 1-16, 17-32, 33-48 and 49-64.

Selecting Sequencer Note Track steps 1-16, 17-32, 33-48 and 49 to 64 / + Controller tracks steps 65-80, 81-96, 97-112, 113-128:



Push and release the **Func/Mute** button, so that it is lightning up. Now, push and release step button 1, **Bar Sel**:



Now step button 1, **Bar Sel**, will light up.

Note tracks: When you are located on any other pages, than the sequencer controller tracks pages, bar is selected to be viewed for a note track. The selected bar, is the bar that is shown on the step buttons, and on the note track pages.

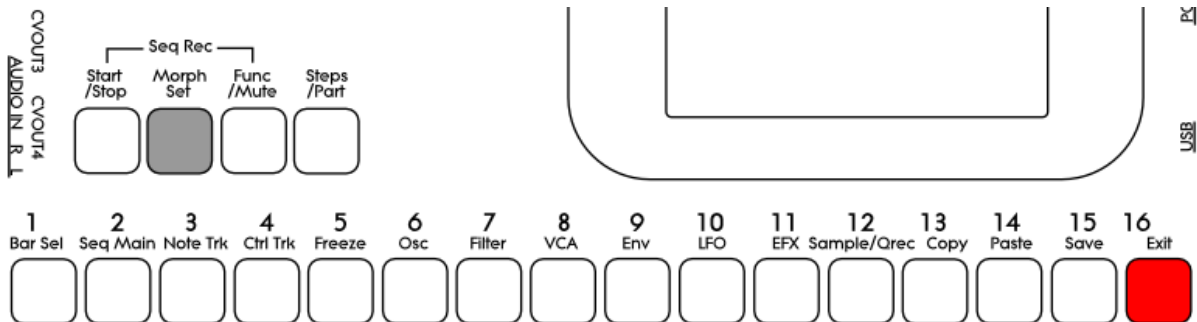
Hit Bar Sel, so that it lights up. Now step buttons 2 to 5 will select bar 1 to 4, of the currently selected track. The button of the bar that is selected, will light up. The other 3 bar select buttons will flash.

Controller tracks: When you are located on any of the controller track pages, bar is selected to be viewed for the currently selected controller track. The selected bar, is the bar that is shown on the controller track pages.

Hit Bar Sel, so that it lights up. Now step buttons 2 to 9 will select bar 1 to 8, of the currently selected track. The button of the bar that is selected, will light up. The other 7 bar select buttons will flash.

When pushing a bar button, that is already selected, both on note and controller tracks, the function of the Edit Knobs will now be toggled: Parameters > Knobs1-8 > Knobs 9-16.

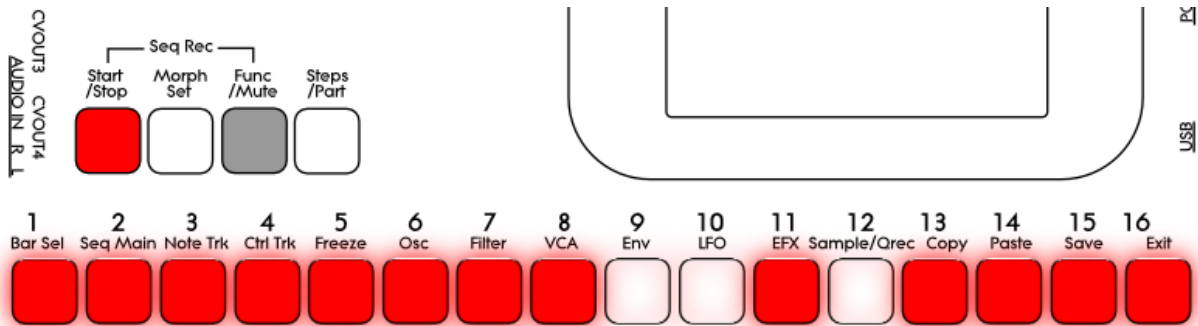
Setting the last step of a Note Track



This can be done on the Note Tracks edit pages, but it can also be done via the pushbuttons.

Push and hold the Morph Set button. If the last step of the selected note track are inside the selected step range, this will now be shown, as a step button that lights up. Push any step button, while still holding down the Morph Set button, to set the last step of the selected Note Track.

Mute/Unmute Note Tracks:



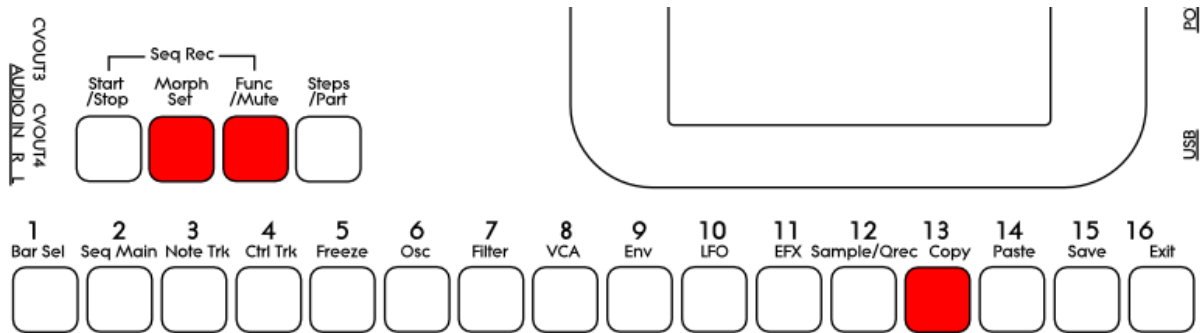
To mute, unmute or view the mute state of the 16 note tracks, push and hold the Func/Mute button.

Unmuted tracks will now be shown by a step button that is lighting up, and shortly flashes off, every time the track is triggering.

Muted tracks are shown by a step button that is unlit, and that lights up shortly, every time the track would have triggered something, if it weren't muted.

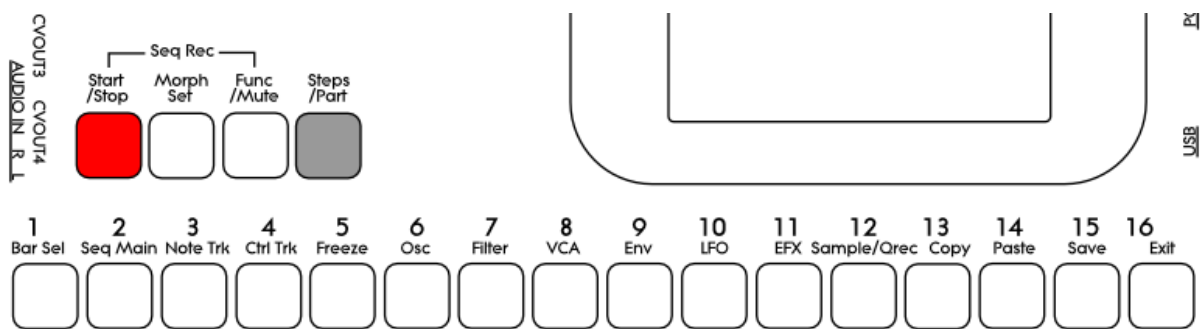
To mute or unmute a track, simply hit the corresponding step button, while still holding down the Func/Mute button.

Copy Morph Layer A to Layer B



Push and release the Func/Mute button, so that it lights up, to enter the function buttons. Now push and release step button 13 (Copy), so that this also lights up. Instructions for copy and panic will now be shown on the screen. Push and release the Morph Set button. Morph layer A synth parts and sequencer parameters has now been copied to morph layer B. If you turn the Morph knobs, you should now hear the same sound/sequence, no matter what position the knobs are in.

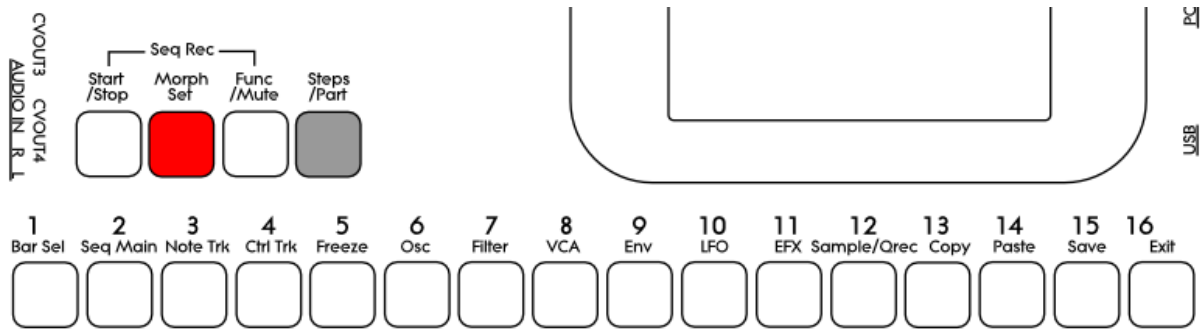
Selecting the next preset



Push and hold the Steps/Part button while pressing the Start/Stop button. LD3 will now select the next preset. If the sequencer is playing back, the Start/Stop LED will now start to flash, and the text "NEXT:" will show right above the new presets name, awaiting track 1 to reach its start/end step. As soon as this happens, LD3 will switch to the newly selected preset, the Start/Stop LED will stop flashing, and "NEXT:" will disappear.

If the sequencer is not playing back, LD3 will immediately switch to the new preset.

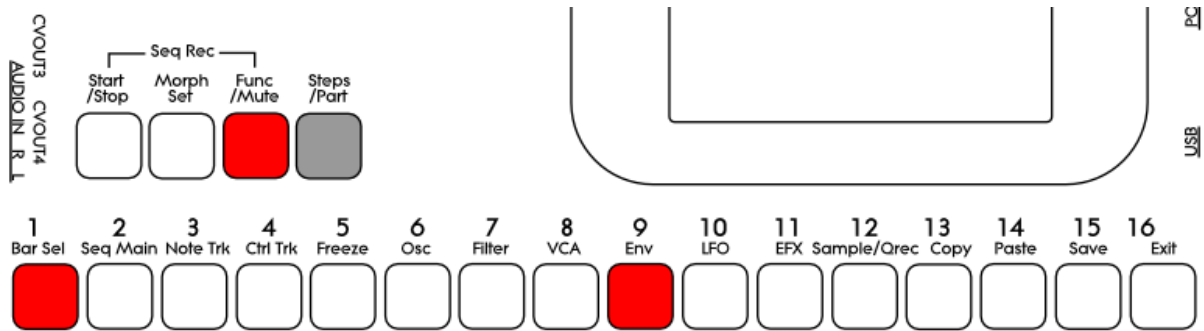
Selecting the previous preset



Push and hold the Steps/Part button while pressing the Morph/Set button. LD3 will now select the previous preset. If the sequencer is playing back, the Start/Stop LED will now start to flash, and the text “NEXT:” will show right above the new presets name, awaiting track 1 to reach its start/end step. As soon as this happens, LD3 will switch to the newly selected preset, the Start/Stop LED will stop flashing, and “NEXT:” will disappear.

If the sequencer is not playing back, LD3 will immediately switch to the new preset.

Selecting Presets using the step buttons



Push and hold the Steps/Part button while pressing the Func/Mute button. Now both of these buttons will light up.

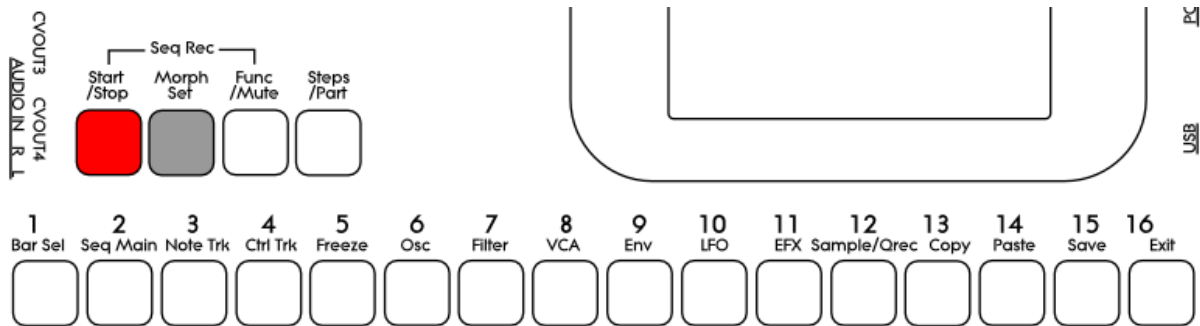
It is now possible to select the 64 presets in the currently selected preset bank, using the step buttons. Step button 1 to 8 selects the lower digit of the preset number, and step button 9-16 selects the higher digit.

When pushing the same button combination again, the Steps/Part button will still light up, while the Func/Mute button will start to flash.

It is now possible to select preset bank A to P, using step buttons 1 to 16.

Pushing the same button combination yet another time, will exit from the preset select mode.

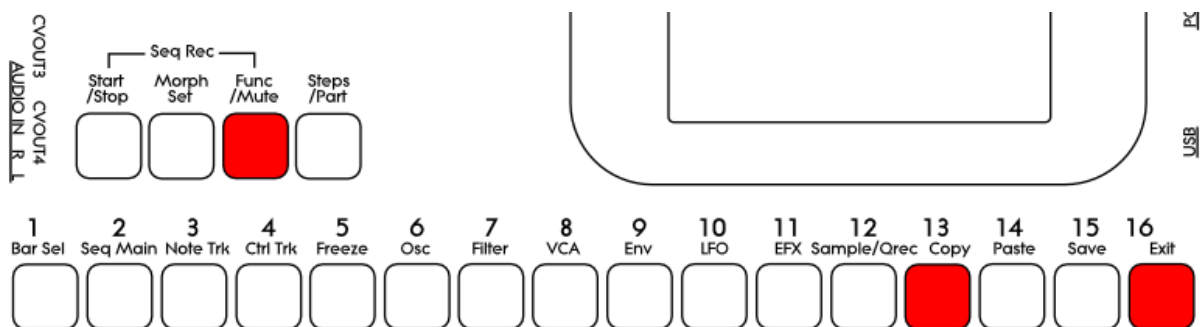
Preset Reload / Advance Song realtime record step



Push and hold the Morph Set button, while pressing the Start/Stop button, to reload the currently selected preset.

While realtime recording a song, reloading the preset, will cause the song recorder to advance to the next song step. This is useful for recording track mutes of the same preset into the song.

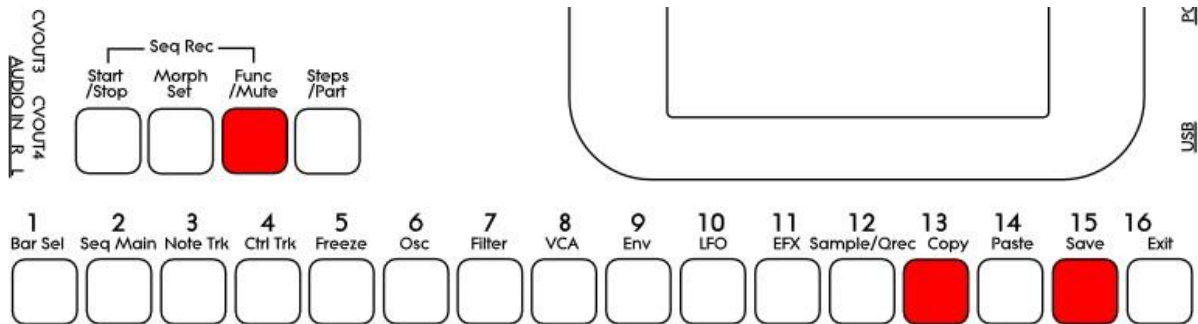
Panic



Push and release the Func/Mute button, so that it lights up, to enter the function buttons. Now push and release step button 13 (Copy), so that this also lights up. Instructions for copy and panic will now be shown on the screen. Push and release step button 16 (Exit).

Now all notes, both internally and on any MIDI devices connected to the LD3 MIDI out, will be shutted off.

Compare Edited Preset With Saved Preset



When you are editing a preset, and you would like to compare this with the originally saved preset, this is possible, using the Compare function.

To listen to the previously saved preset:

- Push and release the Func/Mute button, so that it lights up.
- Push and release step button 13 (Copy).
- Push and release step button 15 (Save).

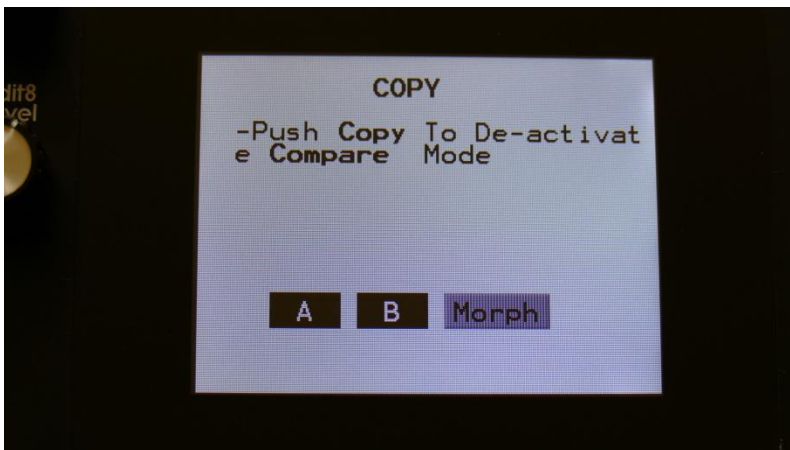
You can now play with the previously saved preset.

To de-activate the Compare function, and jump back to the edited preset:

- Hit step button 15 (Save) again.

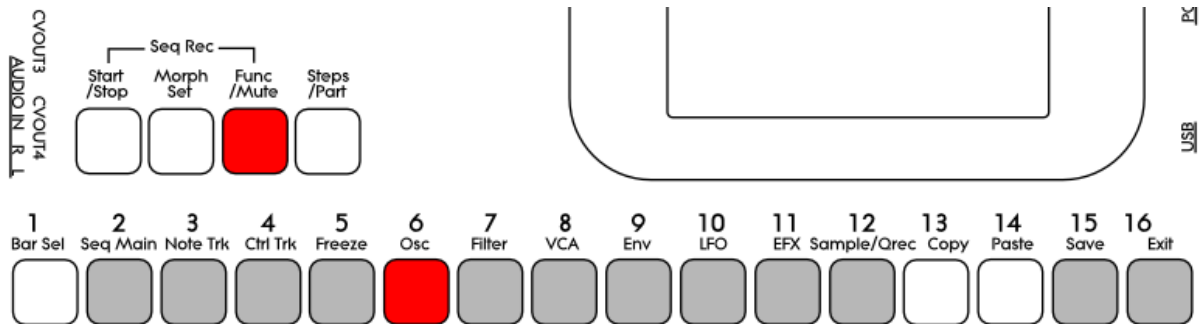
To de-activate the Compare function, and discard your edits:

- Hit step button 13 (Copy). The display will now show:



If you are absolutely sure, that you would like to discard your edits, hit step button 13 (Copy) again.

Shortcuts to Edit Pages



Push and release the Func/Mute button, so that it lights up. Now push and release the step button, that has the name of the edit page over it, that you wish to jump to. When pushing some of these buttons a number of times, it will toggle through a number of pages:

Seq Main: Sequencer Main.

Note Trk: Notes > Gate time > Velocity > Note Mod.

Ctrl Trk: Ctrl Track 1-16 step values > Ctrl Track 1-16 CC > Ctrl Track 17-32 step values > Ctrl Track 17-32 CC.

Freeze: Freezes any effects, that has Freeze modulation set to “Buton”.

Osc: Osc > Mod > Sel > Smp.

Filter: Analog Filters > Digital filter.

VCA: VCA > Mod.

Env: Envelope > Mod.

LFO: LFO > Mod.

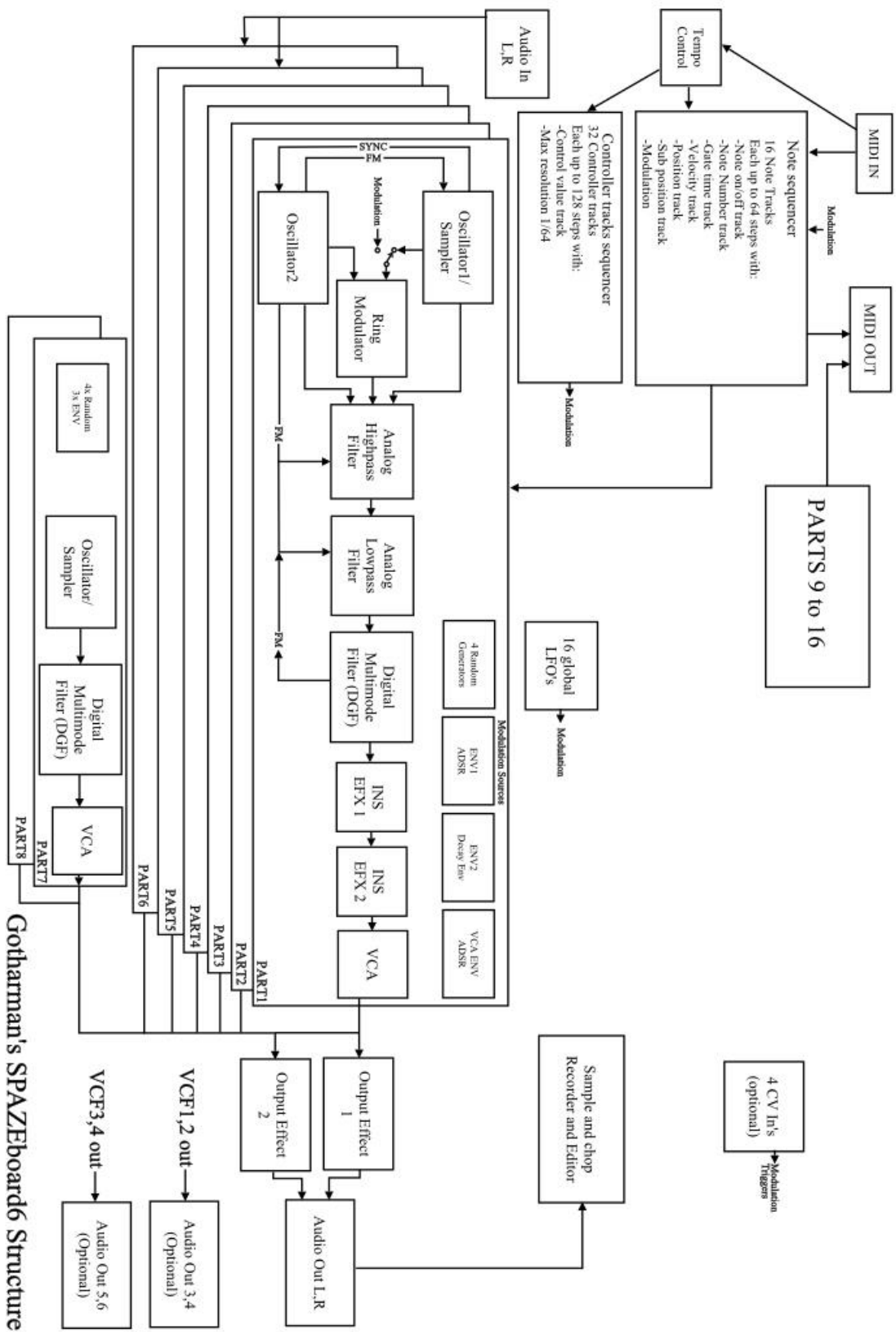
EFX: EFX Select > EFX > Mod.

Sample/Qrec: Sample Rec > Starts Quick recording > Stops Quick Recording and saves sampling.

Sample/Qrec and then Exit: Sample Edit.

Save: Save To Preset select > Save to name 1-8 > Save to name 9-16 > Saving Preset.

Exit: Exits from any page.



Gotharman's SPAZKeyboard6 Structure

The Synth/Sampler Parts

Spazeboard6 has 16 parts:

-**Part 1 to 6** has 2 oscillators (1 can play back samples), a ring modulator, 2 resonant analog filters (HPF + LPF), a resonant digital multimode filter, 2 insert effects, 1 VCA, 3 envelopes, 4 key-triggered random generators and 2 granular modulators.

-**Part 7 and 8** has 1 oscillator (with sample playback), 1 resonant digital multimode filter, 1 VCA, 3 envelopes, 4 key-triggered random generators and 2 granular modulators.

-**Part 9 to 16** is for controlling external gear, and has no synth parts at all.

Some modulation sources is shared between the first 8 parts:

-**16 LFO's** with morphable waveforms.

-**8 LFO triggered random generators.**

-**2 Output Effect Processors** are available just before the stereo outputs. Each of the 8 first parts can be sent to these.

Remember to save all edits you do in the synth/sampler section. Else they will be lost when you change preset, or turn LD3 off. See how to in the "Save Preset" section.

List of Modulation Sources:

Env1: The output of ADSR Envelope 1

Env1-: The output of ADSR Envelope 1 Inverted

Env2: The output of Decay Envelope 2

Env2-: The output of Decay Envelope 2 Inverted

Aenv: The output of the VCA Envelope

Aenv-: The output of the VCA Envelope Inverted

LFO1: The output of LFO1

LFO1-: The output of LFO1 Inverted

LFO2: The output of LFO2

LFO2-: The output of LFO2 Inverted

LFO3: The output of LFO3

LFO3-: The output of LFO3 Inverted

LFO4: The output of LFO4

LFO4-: The output of LFO4 Inverted

LFO5: The output of LFO5

LFO5-: The output of LFO5 Inverted

LFO6: The output of LFO6

LFO6-: The output of LFO6 Inverted

LFO7: The output of LFO7

LFO7-: The output of LFO7 Inverted

LFO8: The output of LFO8

LFO8-: The output of LFO8 Inverted

LFO9: The output of LFO9

LFO9-: The output of LFO9 Inverted

LFO10: The output of LFO10

LFO10-: The output of LFO10 Inverted

LFO11: The output of LFO11

LFO11-: The output of LFO11 Inverted

LFO12: The output of LFO12

LFO12-: The output of LFO12 Inverted

LFO13: The output of LFO13

LFO13-: The output of LFO13 Inverted

LFO14: The output of LFO14

LFO14-: The output of LFO14 Inverted

LFO15: The output of LFO15

LFO15-: The output of LFO15 Inverted

LFO16: The output of LFO16
LFO16-: The output of LFO16 Inverted
Rnd1: The output of Part Random Generator 1
Rnd1-: The output of Part Random Generator 1 Inverted
Seq1: The output of Sequencer Controller Track 1
Seq1-: The output of Sequencer Controller Track 1 Inverted
Seq2: The output of Sequencer Controller Track 2
Seq2-: The output of Sequencer Controller Track 2 Inverted
Seq3: The output of Sequencer Controller Track 3
Seq3-: The output of Sequencer Controller Track 3 Inverted
Seq4: The output of Sequencer Controller Track 4
Seq4-: The output of Sequencer Controller Track 4 Inverted
Seq5: The output of Sequencer Controller Track 5
Seq5-: The output of Sequencer Controller Track 5 Inverted
Seq6: The output of Sequencer Controller Track 6
Seq6-: The output of Sequencer Controller Track 6 Inverted
Seq7: The output of Sequencer Controller Track 7
Seq7-: The output of Sequencer Controller Track 7 Inverted
Seq8: The output of Sequencer Controller Track 8
Seq8-: The output of Sequencer Controller Track 8 Inverted
Seq9: The output of Sequencer Controller Track 9
Seq9-: The output of Sequencer Controller Track 9 Inverted
Seq10: The output of Sequencer Controller Track 10
Seq10-: The output of Sequencer Controller Track 10 Inverted
Seq11: The output of Sequencer Controller Track 11
Seq11-: The output of Sequencer Controller Track 11 Inverted
Seq12: The output of Sequencer Controller Track 12
Seq12-: The output of Sequencer Controller Track 12 Inverted
Seq13: The output of Sequencer Controller Track 13
Seq13-: The output of Sequencer Controller Track 13 Inverted
Seq14: The output of Sequencer Controller Track 14
Seq14-: The output of Sequencer Controller Track 14 Inverted
Seq15: The output of Sequencer Controller Track 15
Seq15-: The output of Sequencer Controller Track 15 Inverted
Seq16: The output of Sequencer Controller Track 16
Seq16-: The output of Sequencer Controller Track 16 Inverted
Seq17: The output of Sequencer Controller Track 17
Seq17-: The output of Sequencer Controller Track 17 Inverted
Seq18: The output of Sequencer Controller Track 18
Seq18-: The output of Sequencer Controller Track 18 Inverted

Seq19: The output of Sequencer Controller Track 19
Seq19-: The output of Sequencer Controller Track 19 Inverted
Seq20: The output of Sequencer Controller Track 20
Seq20-: The output of Sequencer Controller Track 20 Inverted
Seq21: The output of Sequencer Controller Track 21
Seq21-: The output of Sequencer Controller Track 21 Inverted
Seq22: The output of Sequencer Controller Track 22
Seq22-: The output of Sequencer Controller Track 22 Inverted
Seq23: The output of Sequencer Controller Track 23
Seq23-: The output of Sequencer Controller Track 23 Inverted
Seq24: The output of Sequencer Controller Track 24
Seq24-: The output of Sequencer Controller Track 24 Inverted
Seq25: The output of Sequencer Controller Track 25
Seq25-: The output of Sequencer Controller Track 25 Inverted
Seq26: The output of Sequencer Controller Track 26
Seq26-: The output of Sequencer Controller Track 26 Inverted
Seq27: The output of Sequencer Controller Track 27
Seq27-: The output of Sequencer Controller Track 27 Inverted
Seq28: The output of Sequencer Controller Track 28
Seq28-: The output of Sequencer Controller Track 28 Inverted
Seq29: The output of Sequencer Controller Track 29
Seq29-: The output of Sequencer Controller Track 29 Inverted
Seq30: The output of Sequencer Controller Track 30
Seq30-: The output of Sequencer Controller Track 30 Inverted
Seq31: The output of Sequencer Controller Track 31
Seq31-: The output of Sequencer Controller Track 31 Inverted
Seq32: The output of Sequencer Controller Track 32
Seq32-: The output of Sequencer Controller Track 32 Inverted
Kybd: The last note number value received for the part
Kybd-: The last note number value received for the part Inverted
Velo: The last note velocity value received for the part
Velo-: The last note velocity value received for the part Inverted
Maft: The last mono aftertouch value received for the part
Maft-: The last mono aftertouch value received for the part Inverted
Bnd: The last pitch bend value received for the part
Bnd-: The last pitch bend value received for the part Inverted
Knb1: Edit knob 1 value or the last MIDI CC 4 value received
Knb1-: Edit knob 1 value or the last MIDI CC 4 value received Inverted
Knb2: Edit knob 2 value or the last MIDI CC 5 value received
Knb2-: Edit knob 2 value or the last MIDI CC 5 value received Inverted

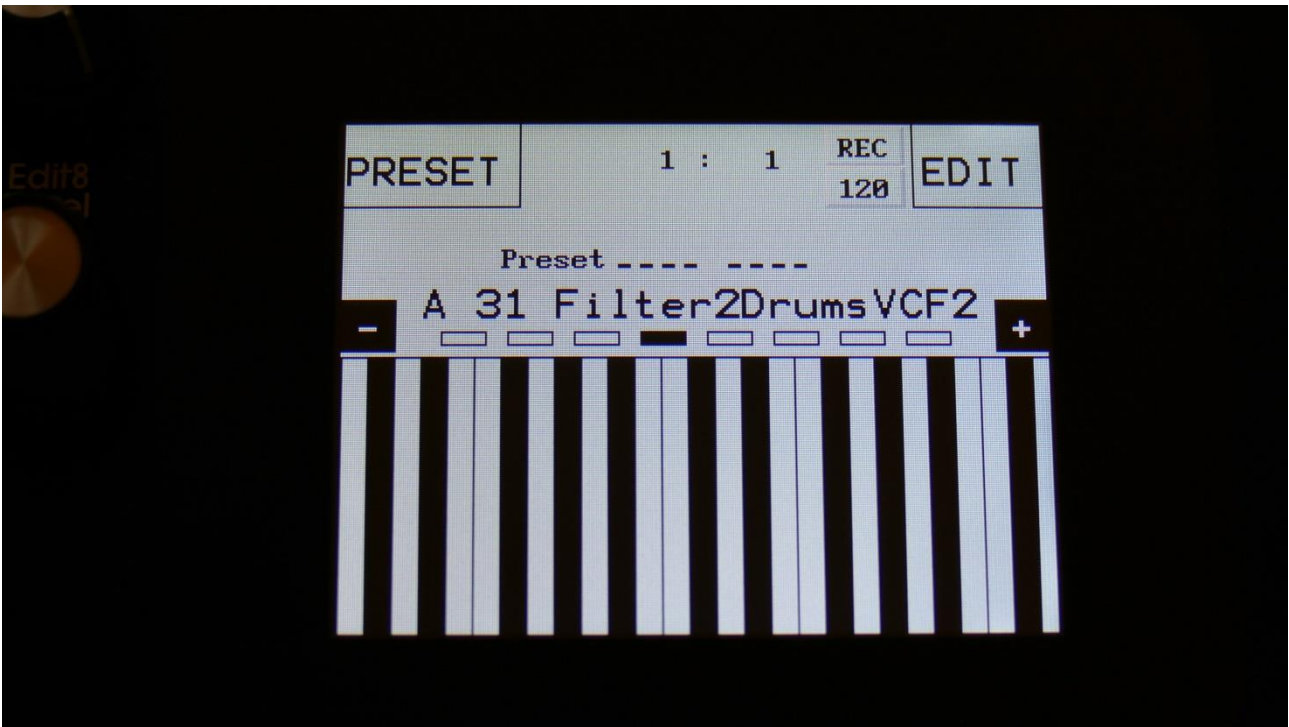
Kn3: Edit knob 3 value or the last MIDI CC 8 value received
Kn3-: Edit knob 3 value or the last MIDI CC 8 value received Inverted
Kn4: Edit knob 4 value or the last MIDI CC 9 value received
Kn4-: Edit knob 4 value or the last MIDI CC 9 value received Inverted
TouY: Touch screen keyboard Y-axis position
TouY-: Touch screen keyboard Y-axis position Inverted
GrRn: Granular Random Modulator
GrRn-: Granular Random Modulator Inverted
GrSq: Granular Sequenced Modulator
GrSq-: Granular Sequenced Modulator Inverted
CV1: The voltage applied to CV Input 1
CV1-: The voltage applied to CV Input 1 Inverted
CV2: The voltage applied to CV Input 2
CV2-: The voltage applied to CV Input 2 Inverted
CV3: The voltage applied to CV Input 3
CV3-: The voltage applied to CV Input 3 Inverted
CV4: The voltage applied to CV Input 4
CV4-: The voltage applied to CV Input 4 Inverted
Lrn1: Random generator triggered by LFO1
Lrn1-: Random generator triggered by LFO1 Inverted
Lrn2: Random generator triggered by LFO2
Lrn2-: Random generator triggered by LFO2 Inverted
Lrn3: Random generator triggered by LFO3
Lrn3-: Random generator triggered by LFO3 Inverted
Lrn4: Random generator triggered by LFO4
Lrn4-: Random generator triggered by LFO4 Inverted
Lrn5: Random generator triggered by LFO5
Lrn5-: Random generator triggered by LFO5 Inverted
Lrn6: Random generator triggered by LFO6
Lrn6-: Random generator triggered by LFO6 Inverted
Lrn7: Random generator triggered by LFO7
Lrn7-: Random generator triggered by LFO7 Inverted
Lrn8: Random generator triggered by LFO8
Lrn8-: Random generator triggered by LFO8 Inverted
Kn5: Edit knob 5 value or the last MIDI CC 10 value received
Kn5-: Edit knob 5 value or the last MIDI CC 10 value received Inverted
Kn6: Edit knob 6 value or the last MIDI CC 11 value received
Kn6-: Edit knob 6 value or the last MIDI CC 11 value received Inverted
Kn7: Edit knob 7 value or the last MIDI CC 12 value received
Kn7-: Edit knob 7 value or the last MIDI CC 12 value received Inverted

Knob8: Edit knob 8 value or the last MIDI CC 7 value received
Knob8-: Edit knob 8 value or the last MIDI CC 7 value received Inverted
Trig: The trigger output of the part
Trig-: The trigger output of the part Inverted
Rnd2: The output of Part Random Generator 2
Rnd2-: The output of Part Random Generator 2 Inverted
Rnd3: The output of Part Random Generator 3
Rnd3-: The output of Part Random Generator 3 Inverted
Rnd4: The output of Part Random Generator 4
Rnd4-: The output of Part Random Generator 4 Inverted
CC17: The last MIDI CC 17 value received
CC17-: The last MIDI CC 17 value received Inverted
CC18: The last MIDI CC 18 value received
CC18-: The last MIDI CC 18 value received Inverted
CC19: The last MIDI CC 19 value received
CC19-: The last MIDI CC 19 value received Inverted
CC20: The last MIDI CC 20 value received
CC20-: The last MIDI CC 20 value received Inverted
CC21: The last MIDI CC 21 value received
CC21-: The last MIDI CC 21 value received Inverted
CC22: The last MIDI CC 22 value received
CC22-: The last MIDI CC 22 value received Inverted
CC23: The last MIDI CC 23 value received
CC23-: The last MIDI CC 23 value received Inverted
CC24: The last MIDI CC 24 value received
CC24-: The last MIDI CC 24 value received Inverted
CC25: The last MIDI CC 25 value received
CC25-: The last MIDI CC 25 value received Inverted
CC26: The last MIDI CC 26 value received
CC26-: The last MIDI CC 26 value received Inverted
CC27: The last MIDI CC 27 value received
CC27-: The last MIDI CC 27 value received Inverted
CC28: The last MIDI CC 28 value received
CC28-: The last MIDI CC 28 value received Inverted
CC29: The last MIDI CC 29 value received
CC29-: The last MIDI CC 29 value received Inverted
CC30: The last MIDI CC 30 value received
CC30-: The last MIDI CC 30 value received Inverted
CC31: The last MIDI CC 31 value received
CC31-: The last MIDI CC 31 value received Inverted

CC33: The last MIDI CC 33 value received
CC33-: The last MIDI CC 33 value received Inverted
CC34: The last MIDI CC 34 value received
CC34-: The last MIDI CC 34 value received Inverted
CC35: The last MIDI CC 35 value received
CC35-: The last MIDI CC 35 value received Inverted
CC36: The last MIDI CC 36 value received
CC36-: The last MIDI CC 36 value received Inverted
CC37: The last MIDI CC 37 value received
CC37-: The last MIDI CC 37 value received Inverted
CC38: The last MIDI CC 38 value received
CC38-: The last MIDI CC 38 value received Inverted
CC39: The last MIDI CC 39 value received
CC39-: The last MIDI CC 39 value received Inverted
CC40: The last MIDI CC 40 value received
CC40-: The last MIDI CC 40 value received Inverted
CC41: The last MIDI CC 41 value received
CC41-: The last MIDI CC 41 value received Inverted
CC42: The last MIDI CC 42 value received
CC42-: The last MIDI CC 42 value received Inverted
CC43: The last MIDI CC 43 value received
CC43-: The last MIDI CC 43 value received Inverted
CC44: The last MIDI CC 44 value received
CC44-: The last MIDI CC 44 value received Inverted
CC45: The last MIDI CC 45 value received
CC45-: The last MIDI CC 45 value received Inverted
CC46: The last MIDI CC 46 value received
CC46-: The last MIDI CC 46 value received Inverted
CC47: The last MIDI CC 47 value received
CC47-: The last MIDI CC 47 value received Inverted
CC48: The last MIDI CC 48 value received
CC48-: The last MIDI CC 48 value received Inverted
CC49: The last MIDI CC 49 value received
CC49-: The last MIDI CC 49 value received Inverted
CC50: The last MIDI CC 50 value received
CC50-: The last MIDI CC 50 value received Inverted
CC51: The last MIDI CC 51 value received
CC51-: The last MIDI CC 51 value received Inverted
CC52: The last MIDI CC 52 value received
CC52-: The last MIDI CC 52 value received Inverted

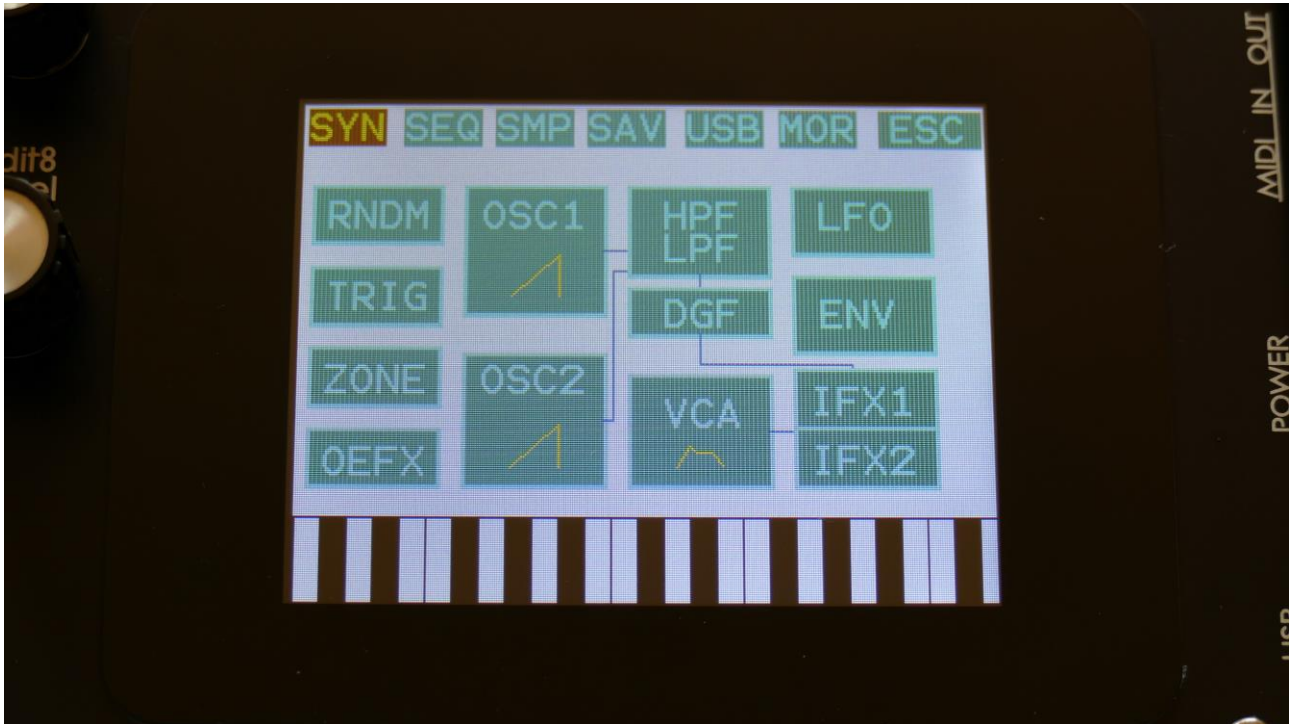
CC53: The last MIDI CC 53 value received
CC53-: The last MIDI CC 53 value received Inverted
CC54: The last MIDI CC 54 value received
CC54-: The last MIDI CC 54 value received Inverted
CC55: The last MIDI CC 55 value received
CC55-: The last MIDI CC 55 value received Inverted
CC56: The last MIDI CC 56 value received
CC56-: The last MIDI CC 56 value received Inverted
CC57: The last MIDI CC 57 value received
CC57-: The last MIDI CC 57 value received Inverted
CC58: The last MIDI CC 58 value received
CC58-: The last MIDI CC 58 value received Inverted
CC59: The last MIDI CC 59 value received
CC59-: The last MIDI CC 59 value received Inverted
CC60: The last MIDI CC 60 value received
CC60-: The last MIDI CC 60 value received Inverted
CC61: The last MIDI CC 61 value received
CC61-: The last MIDI CC 61 value received Inverted

Accessing The Synth Part Pages



From the Preset/Song Select screen, Touch the “EDIT” field.

Now LD3 will show the main Synth page:



In the top of the main Synth page, you will find the 6 main edit groups and the ESC (escape) touch button. Touch any of these group buttons to access them, and touch ESC, to exit to the Preset Select page.

The touch button of the currently selected edit group is brown/yellow, while the buttons of the other groups are green.

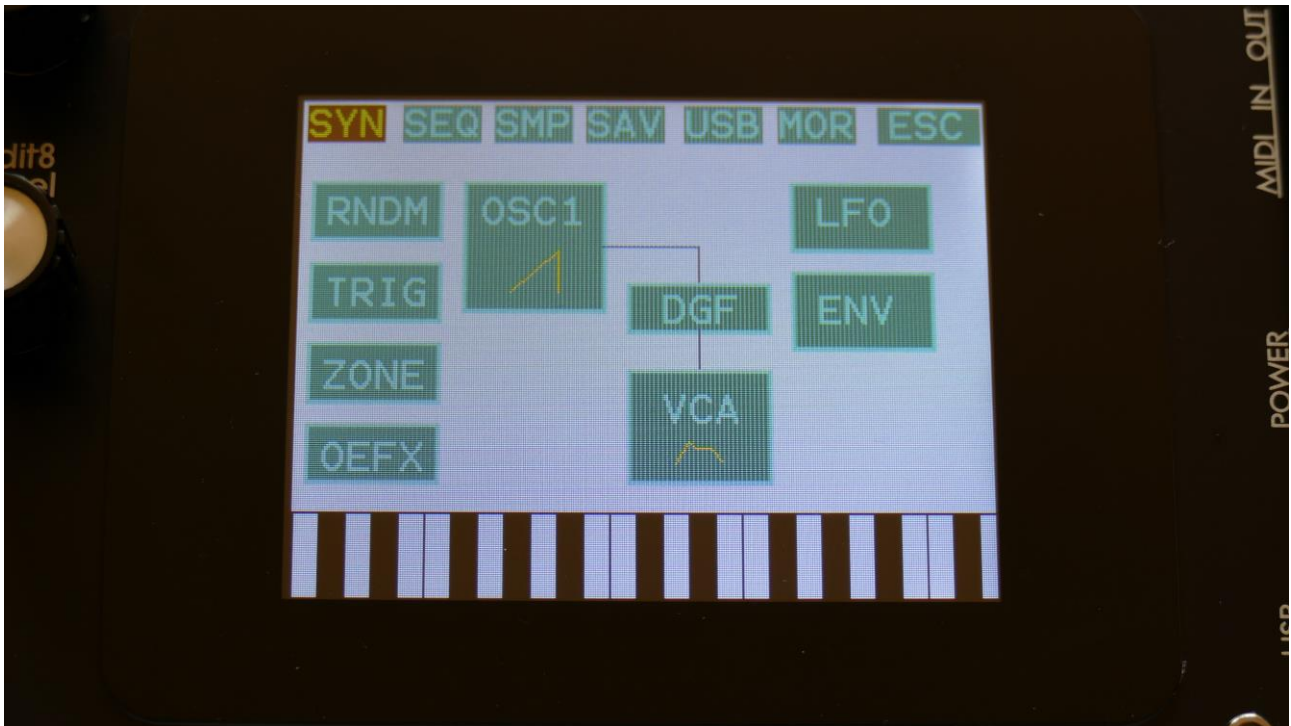
The group of touch buttons, will be referred to as the “group select bar”.

Below the group select bar, you will find the synth blocks. Touch any block, to access the parameters of it, and edit these. Part 1 to 16 is selected using the Steps/Part button in combination with the step buttons.

In the bottom of this page, the touch keyboard is located.

The Synth Part pages can also be accessed, using pushbutton shortcuts. This is explained on page 36 of this manual.

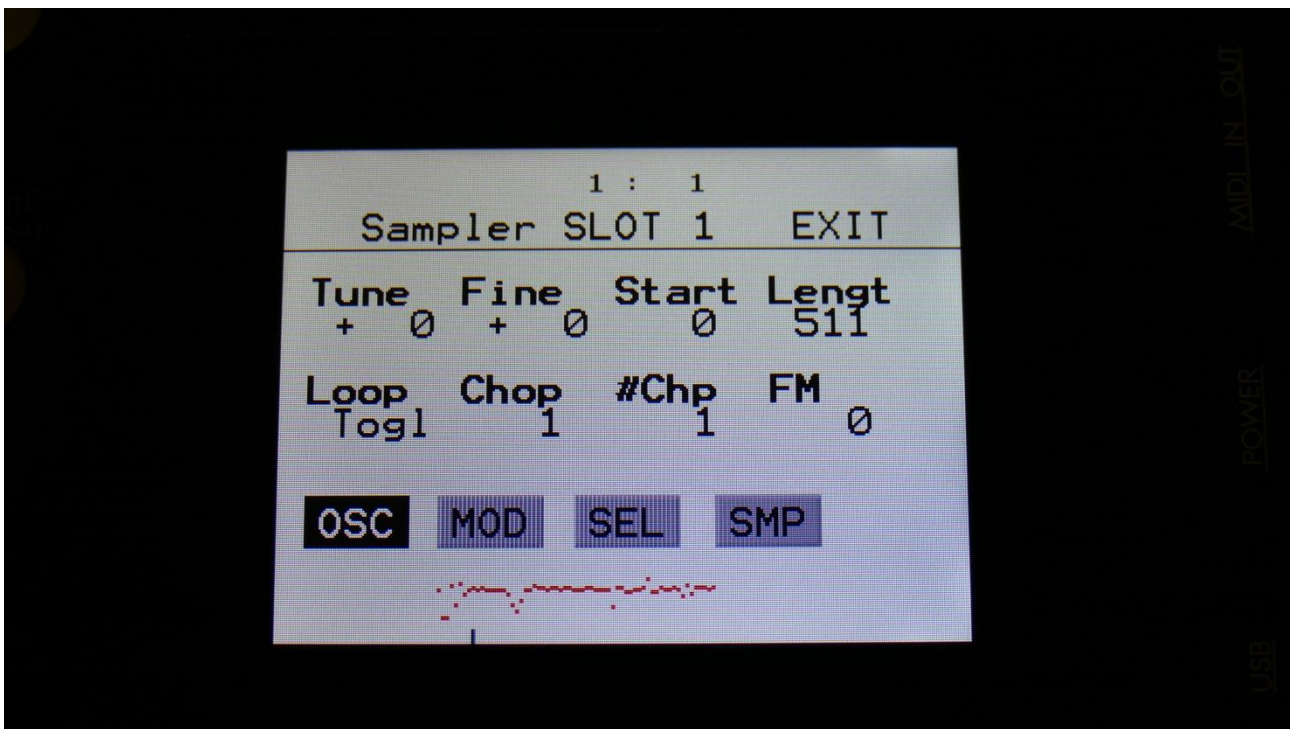
For part 7 and 8, the Synth main page looks like this:



And for part 9 to 16, it looks like this:



Editing The Parameters Of The Synth Part Blocks



Each edit page has up to 8 parameters, that can be edited. The parameters are shown on the display as 8 parameter names, each with an alphanumeric value below them, that shows the current value of the parameter.

When turning any of the 8 Edit Knobs, the corresponding parameter will be adjusted, and you will hear a change in the sound, if the block is active.

Right below the parameters, you will find the subpage select touch buttons. Touch any of these, to access the desired parameter sub page. The touch button that is black with white text, shows that this is the currently selected sub page. The grey touch buttons, are the sub pages, that you can select.

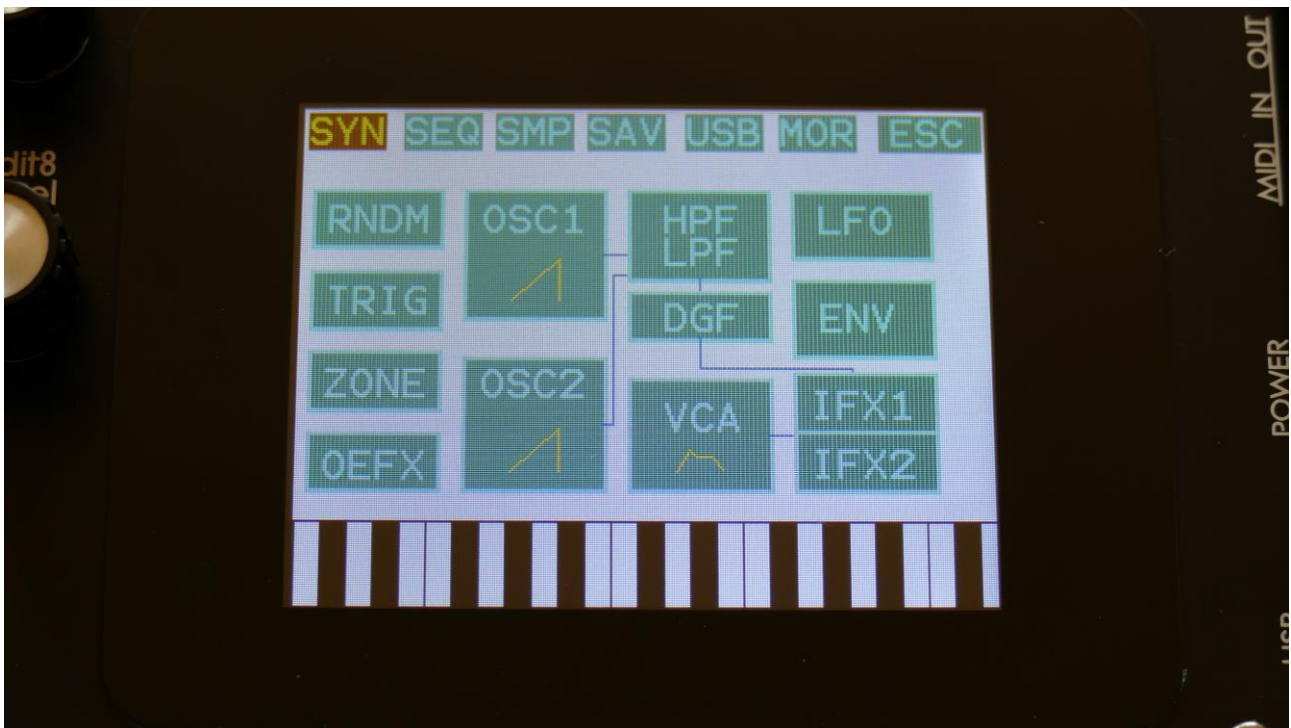
In the top of the display, it is, on the part edit pages, possible to see what part slot that is currently selected.

In the upper right corner of each block you will find "EXIT". Touch this to exit to the main Synth page.

The Synth Part Blocks

In this section of the manual, you will find a description of the parameters of each part, and the parameters related to the LFO's the effects processors and the BUS system.

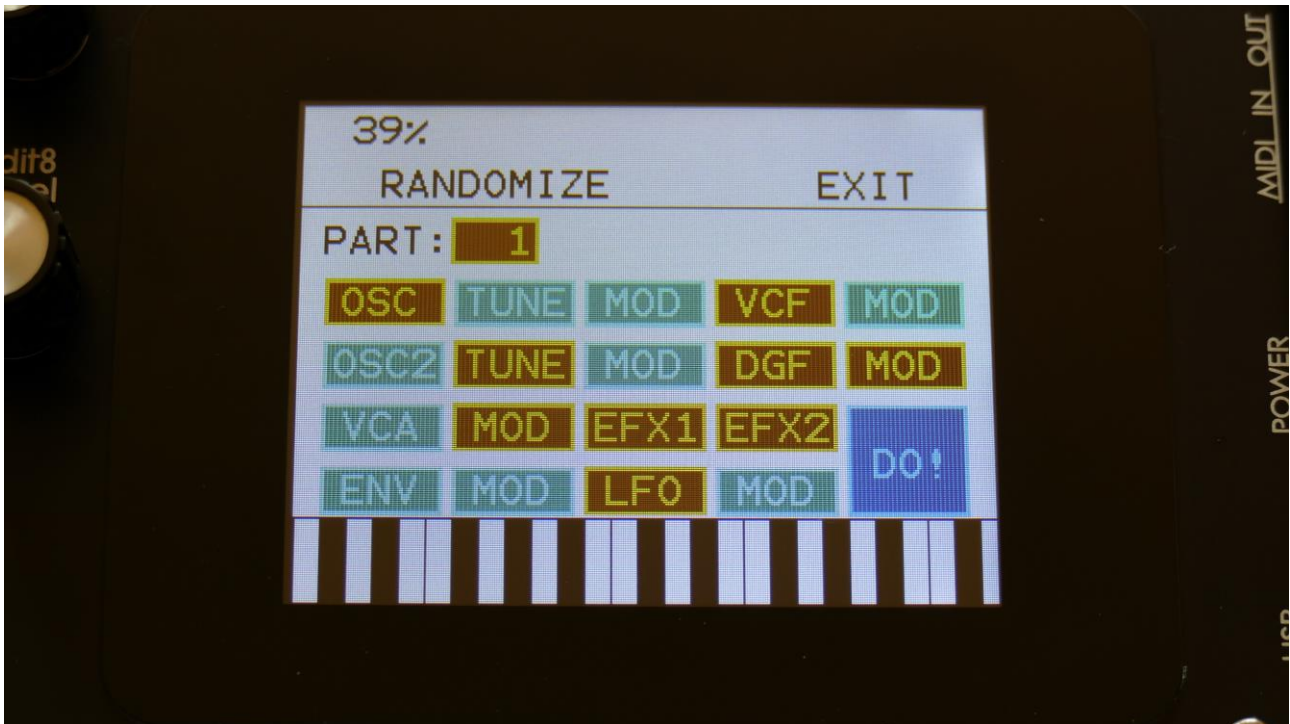
The Synth Part Parameters Randomizer



If you should ever need some new inspiration for sounds, or if you just want to surprise yourself with some sounds that you never even imagined, the LD3 parameter randomizer might be exactly what you need.

To enter this, from the synth parts main page, touch RNDM.

You should now enter this page:



Select the part that you would like to randomize, in the same way as you would usually select a part, by pushing and holding the Steps/Part button, while pressing one of the first 8 step buttons. Switch the blocks on, that you would like to randomize, simply by touching these, adjust the percentage, that it must maximum change the parameters, using Edit Knob 1, and hit “DO!”.

Green blocks are not randomized, brown blocks are.

Listen to the result.

If you like it, exit the randomizer page, and save the preset.

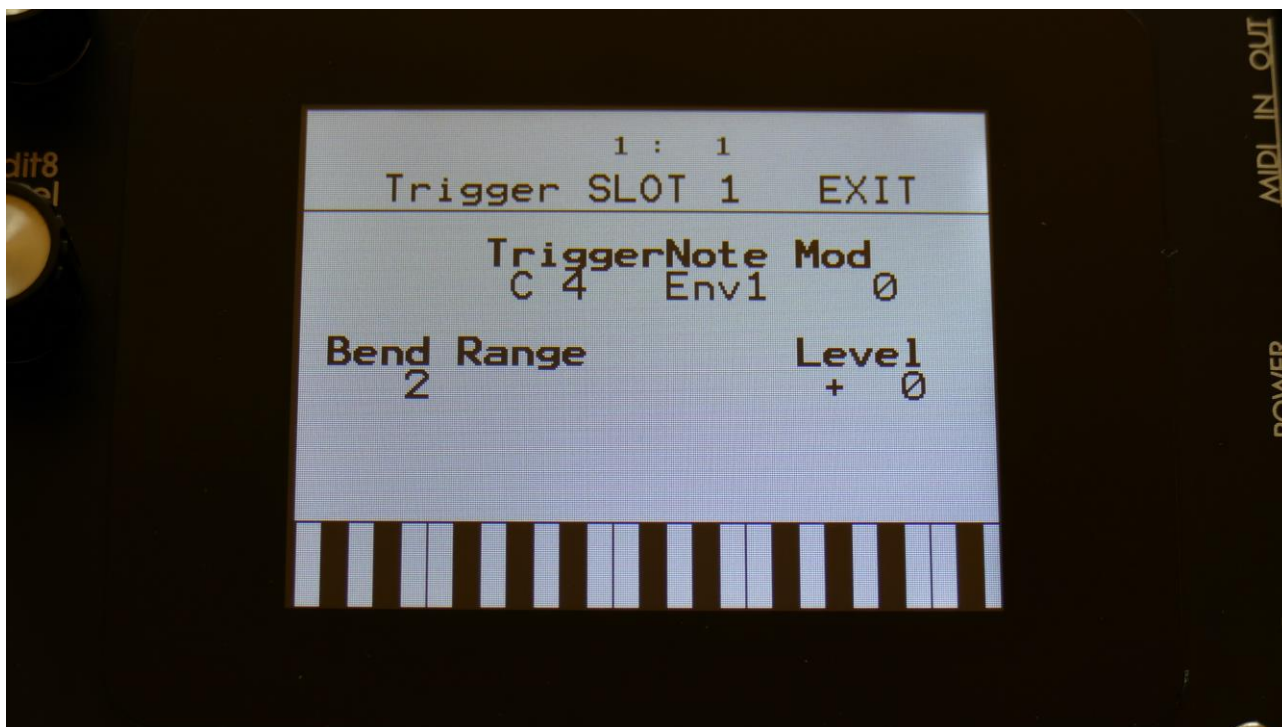
If you don't like the result, hit “DO!” again, and keep hitting it, until something comes up, that you like. Try with different percentage settings, and try to switch different blocks on and off.

When the Randomizer page is entered, all parameters are stored into a temporary buffer, that are used for the randomization. So if you, for instance, first randomizes with 50%, and then with 20%, the result will be maximum 20% away from the initial parameter settings, when the Randomizer page was entered. It does not first randomize 50%, and then randomize 20% on top of that. To randomize things further away, you must exit the Randomizer page, and re-enter it.

Synth Part Trigger Setup for Part 1 to 8

In this block, you can set up the pitch bend range, the note that will play back, when pushing a trigger button, and the overall preset level, for the current preset.

From the synth parts main page, touch TRIG to enter this page.



TriggerNote: C-1 to G9. The note that will play, when the step buttons are in trigger mode (when both the Func/Mute button and the Steps/Part button are unlit), and you push the trigger button, for this part.

This is also the note, that is sent to the internal synth, when a part is triggered by a CV input.

So it is possible to set one CV input up to trigger the part, and use another CV input to modulate the trigger note, for CV/Gate to MIDI conversion.

Trigger Note Modulation Source:

Any modulation source can modulate the trigger note. Only the positive modulation sources, can be selected by the Edit Knob. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

Mod: 0 to 512. Trigger Note Modulation amount

Bend Range: 0 to 12 notes. The global pitch bend range for all parts.

Level: This adjusts the overall level of the entire preset. Negative values attenuates the level, positive values boosts the level. For making the presets levels match each other, without having to adjust the levels of all the 8 parts.

Synth Part Trigger Setup for Part 9 to 16

Parts 9 to 16 are for controlling external MIDI devices. Therefore the trigger set up screen for these, looks a bit different, than for part 1 to 8. In this block, you can set the note that will play pack, when pushing a trigger button, the external control MIDI channel and the overall preset level, for the current preset.

From the synth parts main page, touch TRIG to enter this page.



TriggerNote: C-1 to G9. The note that will be sent via MIDI out, when the step buttons are in trigger mode (when both the Func/Mute button and the Steps/Part button are unlit), and you push the trigger button, for this part.

This is also the note, that is sent to MIDI out, when a part is triggered by a CV input.

So it is possible to set one CV input up to trigger the part, and use another CV input to modulate the trigger note, for CV/Gate to MIDI conversion.

Trigger Note Modulation Source:

Any modulation source can modulate the trigger note. Only the positive modulation sources, can be selected by the Edit Knob. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

Mod: 0 to 512. Trigger Note Modulation amount.

Chan: 1 to 16. The part output MIDI channel. This will also affect the part Sequencer Note Track. This is the MIDI channel that note data played by the trigger button, the touchscreen keyboard, on an attached MIDI device, and from the part sequencer note track are sent on, to the LD3 MIDI out.

Level: This adjusts the overall level of the entire preset. Negative values attenuates the level, positive values boosts the level. For making the presets levels match each other, without having to adjust the levels of all the 4 parts.

Zone Setup

In this block, it is possible to set up a key zone and a MIDI channel for the part, that will take effect, when LD3 is set in multi-timbral mode, and is controlled from an external MIDI device. How to set LD3 in multi-timbral mode are explained later in this manual, in the MOR>Common section.

From the synth parts main page, touch ZONE to enter this page.



LowK1: C-1 to G9. The lowest key of the part zone.

HiK1: C-1 to G9. The highest key of the part zone.

Trps1: -64 to +63. The incoming MIDI notes, inside the key zone, are transposed up(+) or down(-), by the selected value.

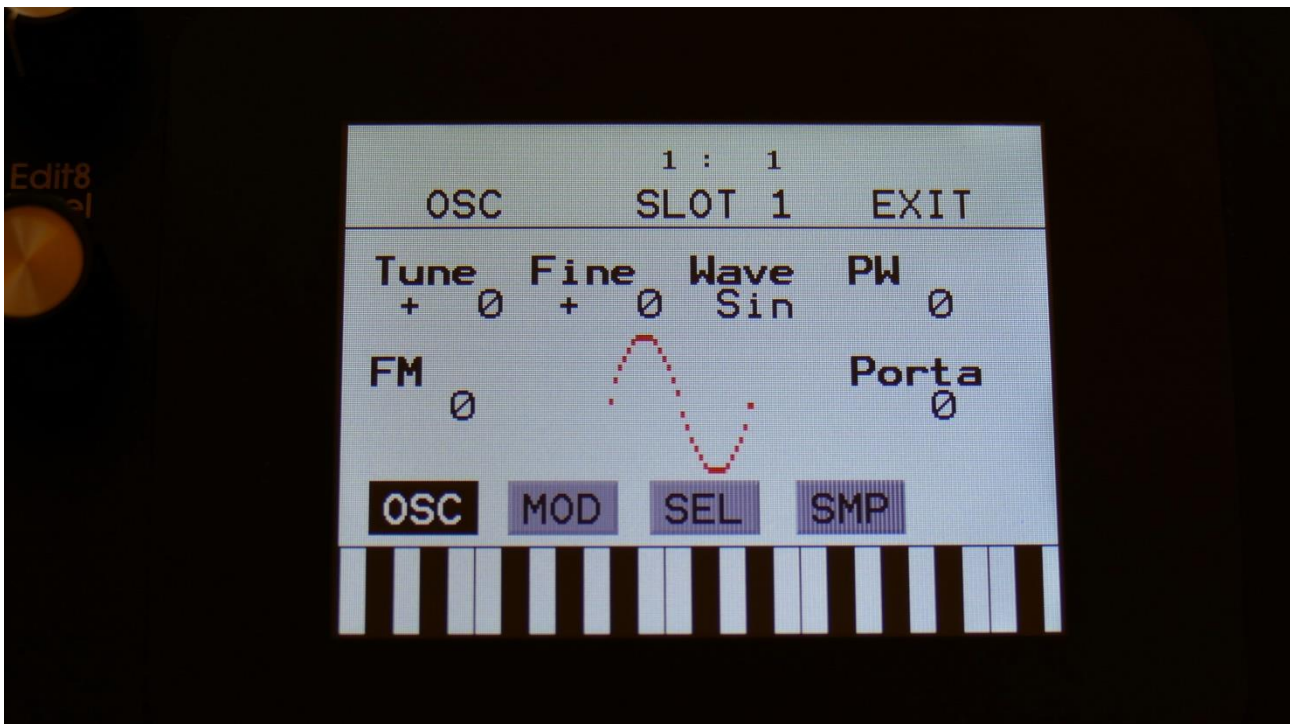
Chan1: 1 to 16. The MIDI channel that will control this part.

Oscillator1/Sampler/Noise Generator

In this block, you will find the settings for the part 1 to 8 oscillator1.

From the synth parts main page, touch OSC1 to enter this page.

OR: Push and release the Func/Mute button, so that it lights up, and push step button 6/Osc 1 time.



The oscillator1 are the first block in the audio chain of a part. This is, together with Oscillator2, responsible for generating the basic sound of each part, that can be modulated, filtered and effected, for shaping a sound.

A Spazeboard6 oscillator1 can be set to act as an oscillator, a sample player or a noise generator.

In oscillator mode it generates a waveform that is morphable between sine, triangle, saw, pulse and feedback waves. Pulse width are adjustable for all waveform types. FM (frequency modulation) is possible, with any audio bus as the modulation source. Pitch, PW, wave and FM amount can be modulated. The pitch range of the oscillators are chromatically over the entire 10 octave MIDI keyboard range.

In sampler mode it plays back any of the 8.191 storable samplings, that can either be recorded on LD3 itself in the Sample Rec section, or be imported in the USB section. Each part sampler has 4

sample slots, that each can contain one sampling. Switching between the 4 samplings is done via the Chop parameter. Pitch, chop, start point and FM amount can be adjusted and modulated, Length can be adjusted. Samples are chromatically tuned, and has a pitch range of 4 octaves above and 5 octaves below the original sample pitch. Loop mode can be set to Off, On, toggle, sustain, off(unchopped) and on(unchopped).

In off and on modes, it will always use the sample chop point, if any.

In toggle mode, the looped sample playback will start, when you push and release a trigger button, and stop when you push and release the same trigger button again.

In sustain mode, it will play the sampling back from the start point, until it reaches the second chop point. Then it will loop back to the first chop point, and keep looping between these 2 points.

When importing a wav sampling with loop points set as cue points, these cue points are imported to the first 2 chop points.

In off and on (unchopped) modes, the whole sampling will play back, ignoring any chop points.

It is possible to create chop points in the LD3 sample editor, and use these. Chops can be detected by level peaks, by single wavecycles, and by dividing the sample length with a settable number.

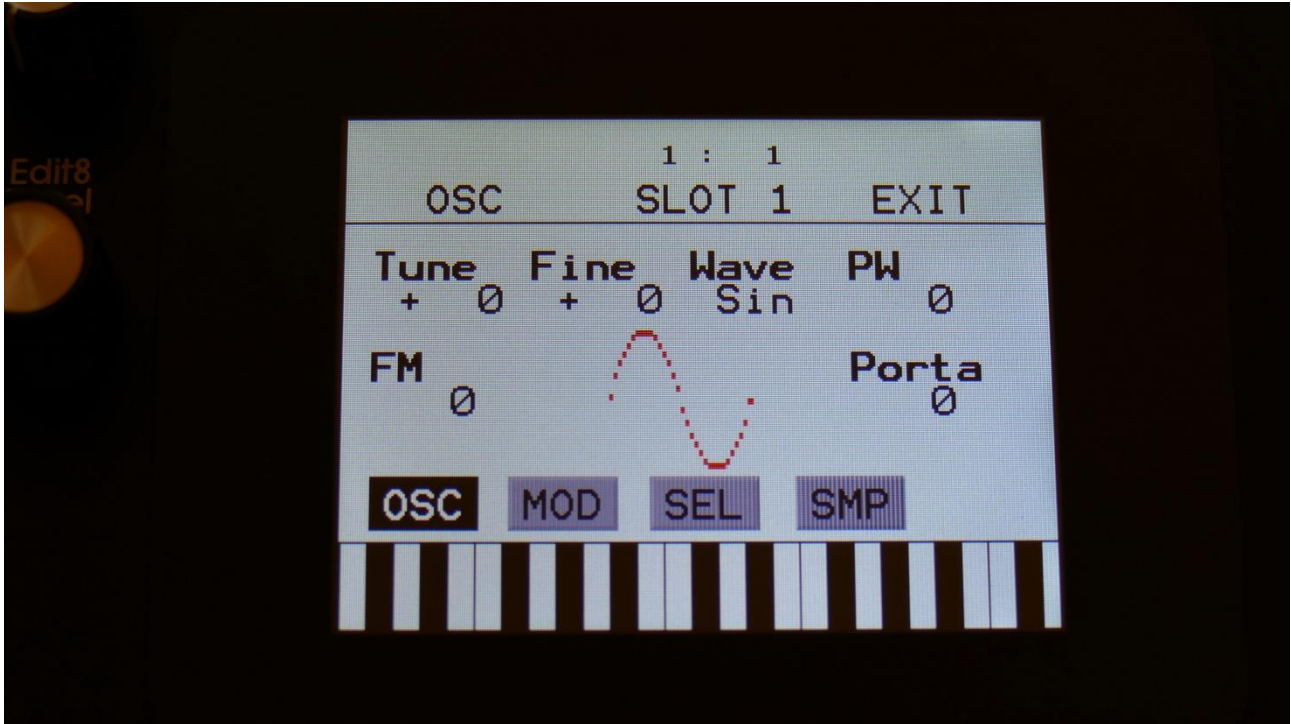
Chop points will also be imported from wav files, containing these as cue points.

Portamento control is provided in both oscillator and sampler mode.

In noise generator mode, the oscillator puts out a noise waveform. Pitch and pw (intensity) can be set and modulated.

Oscillator1 parameters

The OSC page, when in oscillator mode:



In the middle of the oscillator page, the currently generated waveform is shown.

Touch the buttons named OSC, MOD, SEL and SMP, to enter other oscillator pages, like Modulation and oscillator/sampler/noise mode select.

OR: Push and release the Func/Mute button, so that it lights up, and push step button 6/Osc to toggle between these pages.

Tune: Adjust the basic pitch in semitones. Range: -64 to + 63.

Fine: Fine tuning of the pitch. Range: -256 to +255.

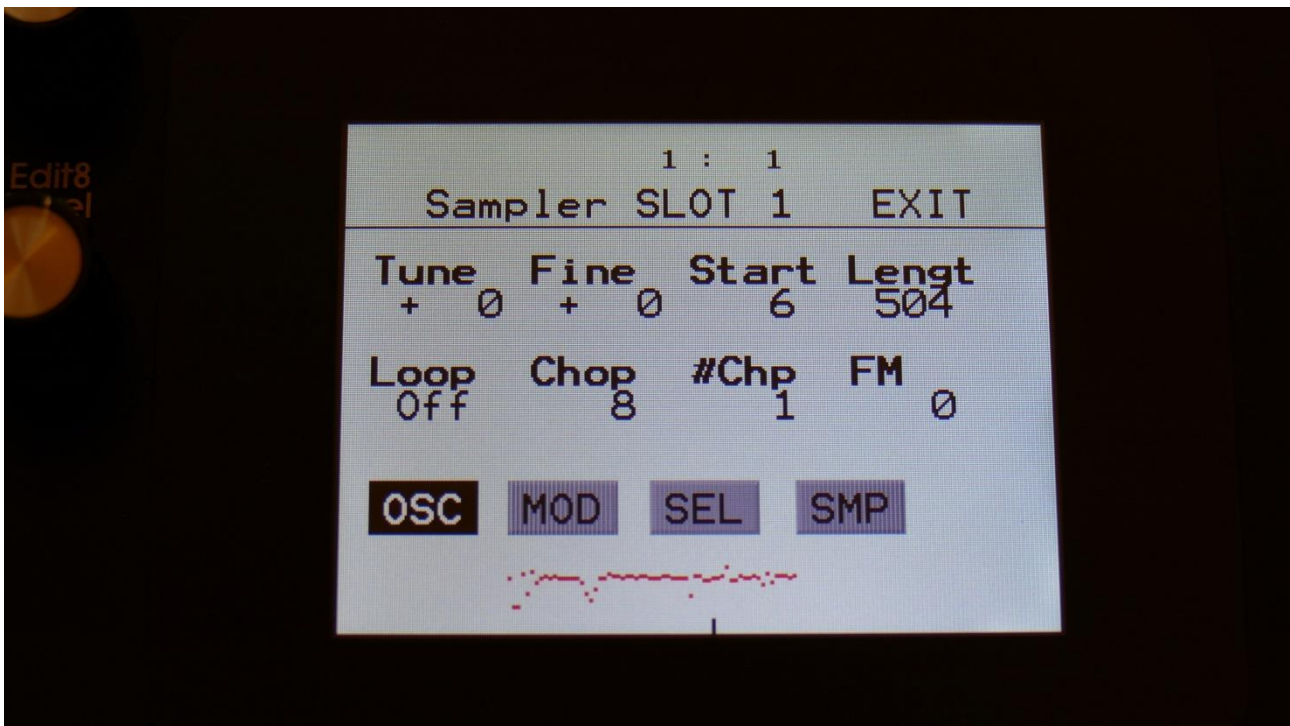
Wave: This parameter lets you morph between sine, triange, saw, pulse and feedback waves.

PW: Adjusts the pulse width of the waveform. Unlike many other oscillator designs, the pulse width can be adjusted on all of LD3's waveforms, not just the pulse wave.

FM: FM amount. The more this is turned up, the more oscillator 2 will modulate the pitch of the oscillator. Range: 0 to 511.

Porta: Portamento. The more this is turned up, the slower the oscillator pitch will slide from one note to another. Range: 0 to 511.

The OSC page, when in Sampler mode:



In the bottom of the sampler page, the selected samplings waveform is shown in rough graphics. When the sample, or part of it, is played back, the small black line below the waveform will show the current playback point.

Tune: Adjust the basic pitch in semitones. Range: -64 to + 63.

Fine: Fine tuning of the pitch. Range: -256 to +255.

Start: The sample start point. Selects at what point the sample will start to play back, when it is triggered. Range: 0 to 511, stretching over the whole sampling.

Length: Sets how much of the sampling should be played back. Range: 0 to 511, stretching over the whole sampling.

Loop: Sets the sampling loop mode.

Off: The sample will not loop, just play back one time from the adjusted, or chop selected, start to end, and then stop. If the selected sampling has chop points, these will always be used.

On: The sample will play back from the adjusted start point, when triggered. When it reaches the adjusted end point, it will loop back to the start point, and play back the sample over and over again. If the selected sampling has chop points, these will always be used for the loop points.

Toggle: The looped sample playback will start, when you push and release a trigger button, and stop when you push and release the same trigger button again.

Sust: When you hit a key, playback of the sampling starts at the sample startpoint.

When the playback reaches the endpoint of the selected chop, it loops back to the chop startpoint, and keeps looping between the chop start and end point, until a key is pressed again.

OfUc: Off, unchopped. The sample will not loop, just play back one time from the adjusted start to end, and then stop. Chop points will be ignored.

OnUc: On, unchopped. The sample will play back from the adjusted start point, when triggered. When it reaches the adjusted end point, it will loop back to the start point, and play back the sample over and over again. Chop points will be ignored.

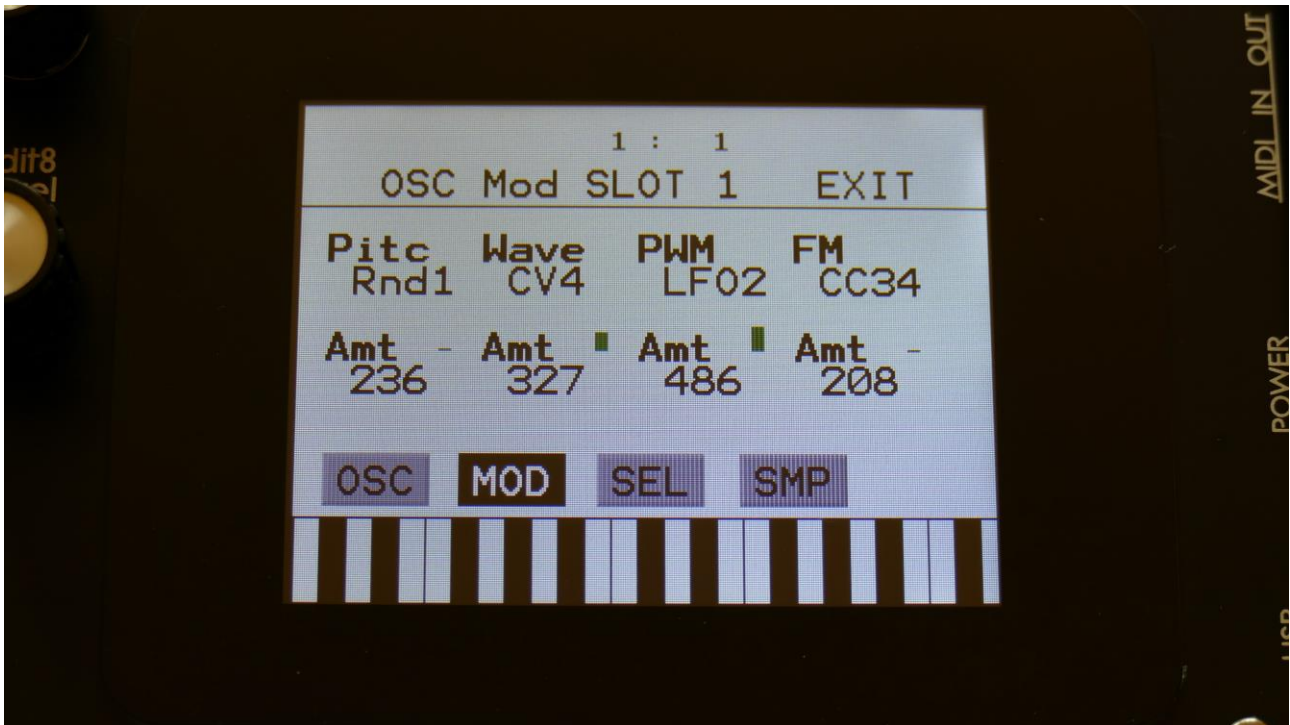
Chop: If chop points has been generated for the selected sampling, a chop can be selected by setting this parameter. If the #Smp parameter on the SEL page is set to 2 or 4 samplings, this parameter will also select which of the 2 or 4 samplings, that will play back, and the chop points for smaple 2 to 3. Range: 0 to 63 for sample 1, 64 to 128 for sample 2, 129 to 192 for sample 3, 193 to 255 for sample 4.

#Chp: The number of Chops to be played back in a row. Range: 1 to 64.

FM: FM amount. The more this is turned up, the more oscillator 2 will modulate the pitch of the sampler. Range: 0 to 511.

Oscillator1 modulation

The MOD page, when in oscillator mode:



The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **The lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Pitc: Modulates the oscillator pitch.

Wave: Modulates the wave select morphing.

PWM: Modulates the pulse width of the waveform

FM: Modulates the FM amount.

The MOD page, when in sampler mode:



The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **The lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Pitc: Modulates the sampler pitch.

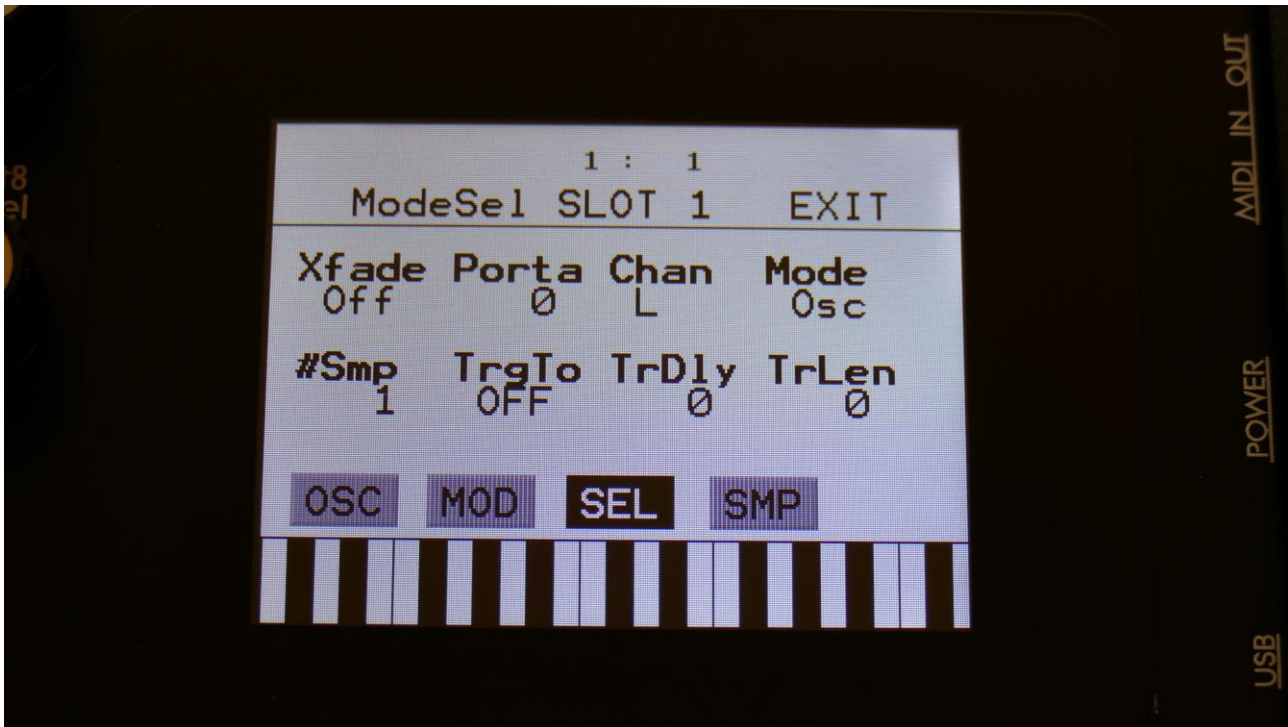
Chop: Modulates the Chop number select, and sample select if #Smp is set to 2 or 4.

Start: Modulates the sample start point.

FM: Modulates the FM amount.

Oscillator/Sampler/Noise mode select

Touch the SEL button, to enter this page. OR: Push and release the Func/Mute button, so that it lights up, and push step button 6/Osc 3 times.



Xfade: Sample fade mode. Only affects sample playback.

Off: No fading are applied to sample playback.

On: When playing back a sampling, both fade in and fade out will be applied.

Especially effective, to obtain click free audio track playback.

Out: When playing back a sampling, fade out will be applied, to remove clicks at the sample end.

Smth: Smoothing. The moment that a sampling loops, the transition from the loop end to the loop start will be smoothed out, to minimize clicks.

Porta: Sampler Portamento. The more this is turned up, the slower the sampler pitch will slide from one note to another. Range: 0 to 511.

Chan: Sets if a stereo sampling should play back the left (L) or the right (R) audio channel.

Mode: Selects whether the part oscillator should act as an oscillator (Osc), a sampler (Smp), or a noise generator (Nois).

#Smp: Number of samplings. Sets if the part should use 1, 2 or 4 samplings. The 4 samplings are selected on the SMP page, and the samplings can be selected, using the Chop parameter on the sampler OSC page.

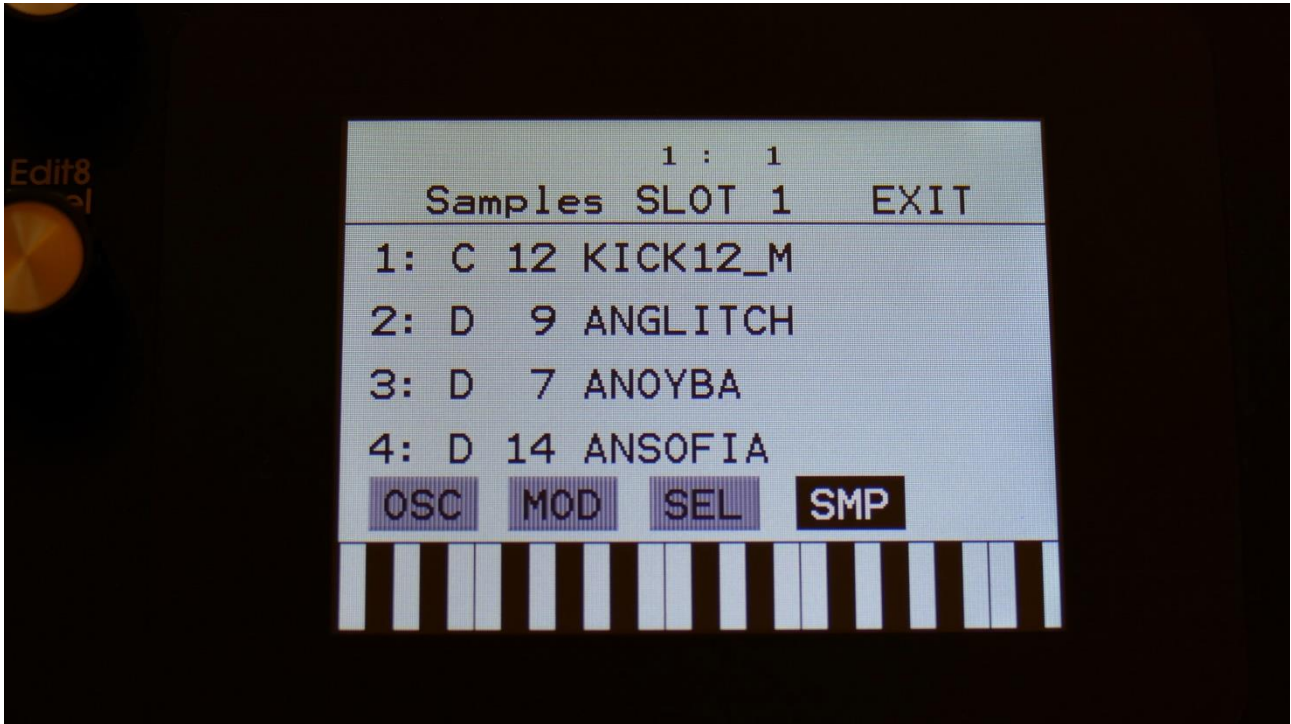
TrgTo: Off, 1 to 16. Makes this part trigger another part. If set to this part, the part will double trigger. The TrLen parameter must be higher than 0, to trigger.

TrDly: 0 to 511. Trigger delay. Will add a delay, before it triggers the other part, if set to any other values than zero.

TrLen: 0 to 511. Trigger length. Sets how long the extra trigger gate should be. If set to zero, it might not trigger.

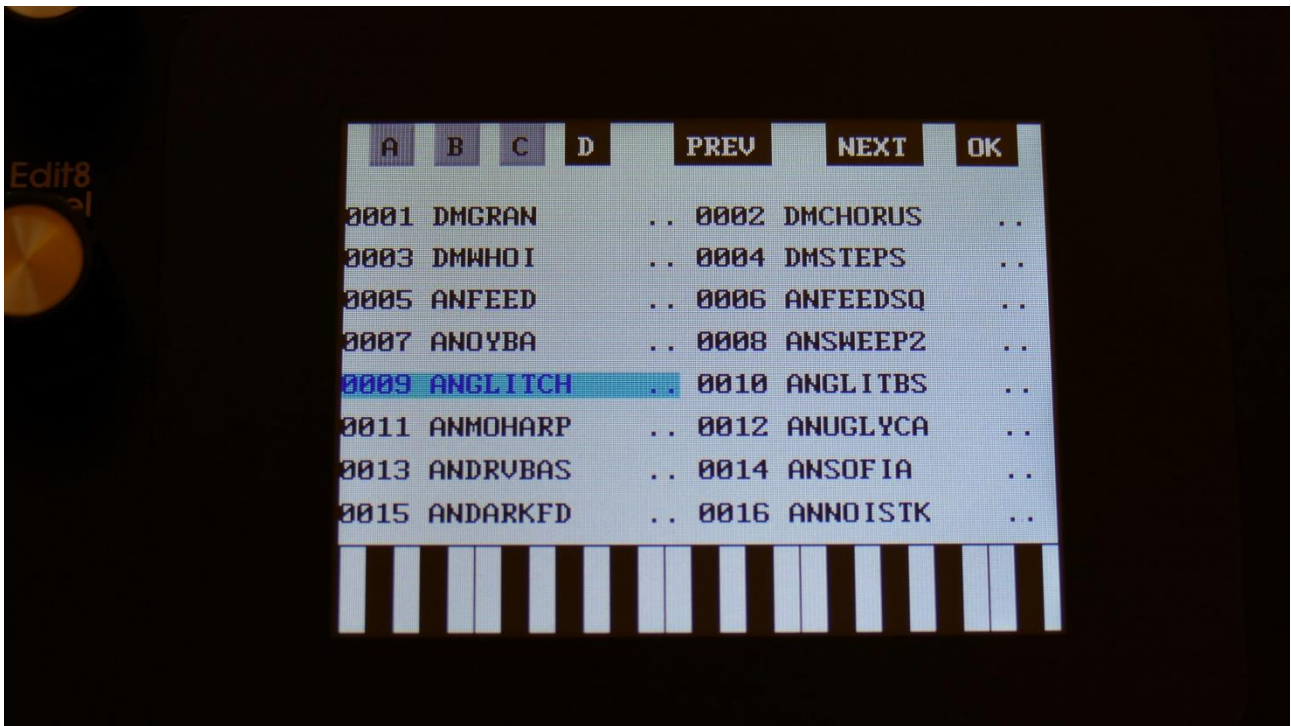
Part Samples Select

Touch the SMP button, to enter this page. OR: Push and release the Func/Mute button, so that it lights up, and push step button 6/Osc 4 times.



Here you can select 4 samplings for the part. The number of these samplings, that are used, is set by the #Smp parameter on the SEL page. The 4 samples can be selected to play back, setting the Chop parameter, and by chop select modulation.

To select a sampling for any of the 4 slots, simply touch the sample name. Now this page will pop up:



On each page you will view 16 of the samplings, that are held in the LD3 FLASH memory, Bank A, B, C or D.

Touch A, B, C and D to select the sample bank, touch PREV and NEXT to view the previous or next 16 samplings.

To select a sampling, touch the sample name.

A small red square is shown near the sample name of the sampling, that are the last one, that was added to the bank.

When you have found the right sampling, touch OK to return to the Synth Part Sample Select page.

The Granular Modulators

Each of the first 8 synth parts of Spazeboard6 has 2 granular modulators, the granular random generator and the granular sequence modulator, that shows as modulation sources, to any parameter that can be modulated.

GrRn – Granular Random Generator

This works in conjunction with the part oscillator, sampler or noise generator.

When in oscillator or noise mode:

Every time the synth waveform starts over, the granular random generator puts out a new random value. When modulating the oscillator wave parameter with this, it will randomly glue different waveforms together, for granular chaos. Since any parameter can be modulated by this, the possibilities are endless.

When in sampler mode:

Every time the sampler loops, the granular random generator will put out a new random value. If you have a wave chopped sampling, and set this parameter up to modulate the chop select parameter, granular random sample chaos will be obtained. Of course, this can also modulate any other parameter, for infinite possibilities.

GrSq – Granular Sequence Modulator

This works in conjunction with the part oscillator, sampler or noise generator, and a Sequencer Controller track. It does not use the output of the controller track directly, only the step values and the last step parameter.

The oscillator of part 1 uses controller track 17 for sequence values, part 2 uses controller track 18, part 3 uses controller track 19 and so on...

When in oscillator or noise mode:

Every time the synth waveform starts over, the granular sequence modulator gets the next value from the associated controller track. When modulating the oscillator wave parameter with this, it is possible to program wave sequences. Since any parameter can be modulated by this, the possibilities are endless.

When in sampler mode:

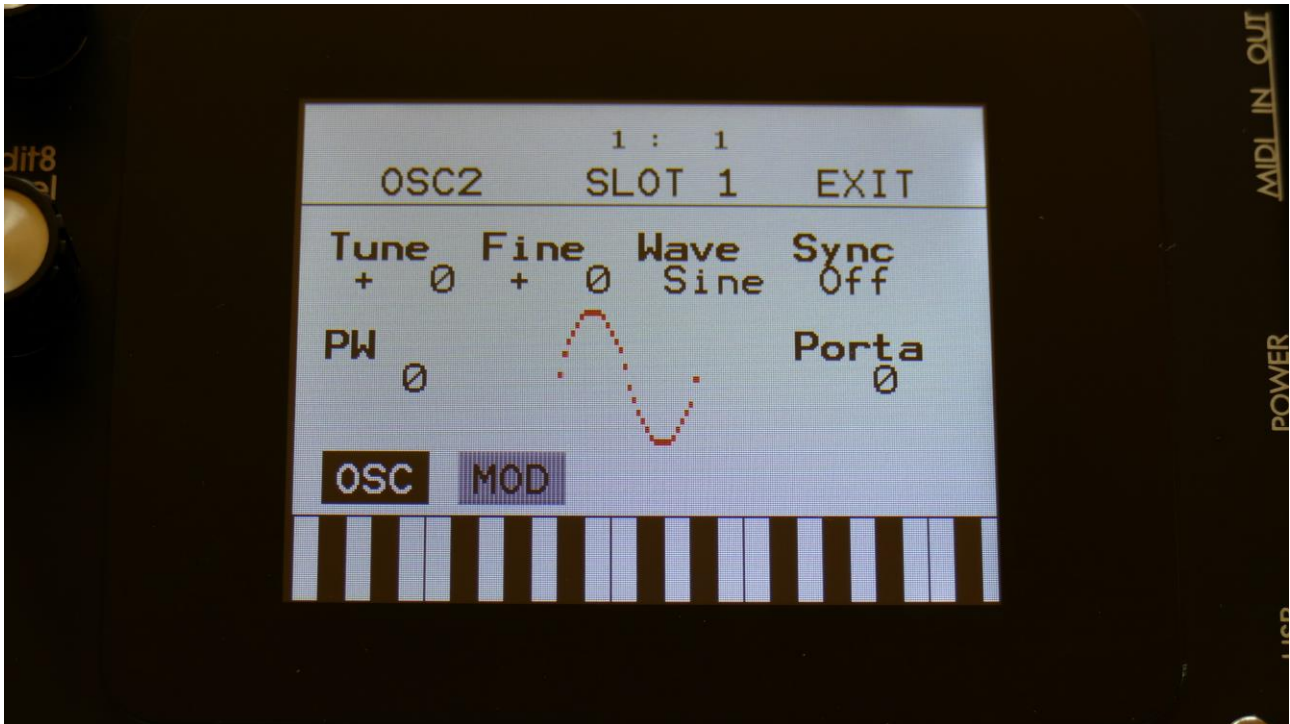
Every time the sampler loops, the granular sequence modulator gets the next value from the associated controller track. If you have a wave chopped sampling, and set this parameter up to modulate the chop select parameter, it is possible to program wave sequences. Of course, this can also modulate any other parameter, for infinite possibilities.

4 Random Generators

Each synth part of LD3 has 4 random generators, that takes in a new random value, every time the part is triggered.

There are no settings to be made for these. They are found as modulation sources, named: Rnd1, Rnd2, Rnd3 and Rnd4.

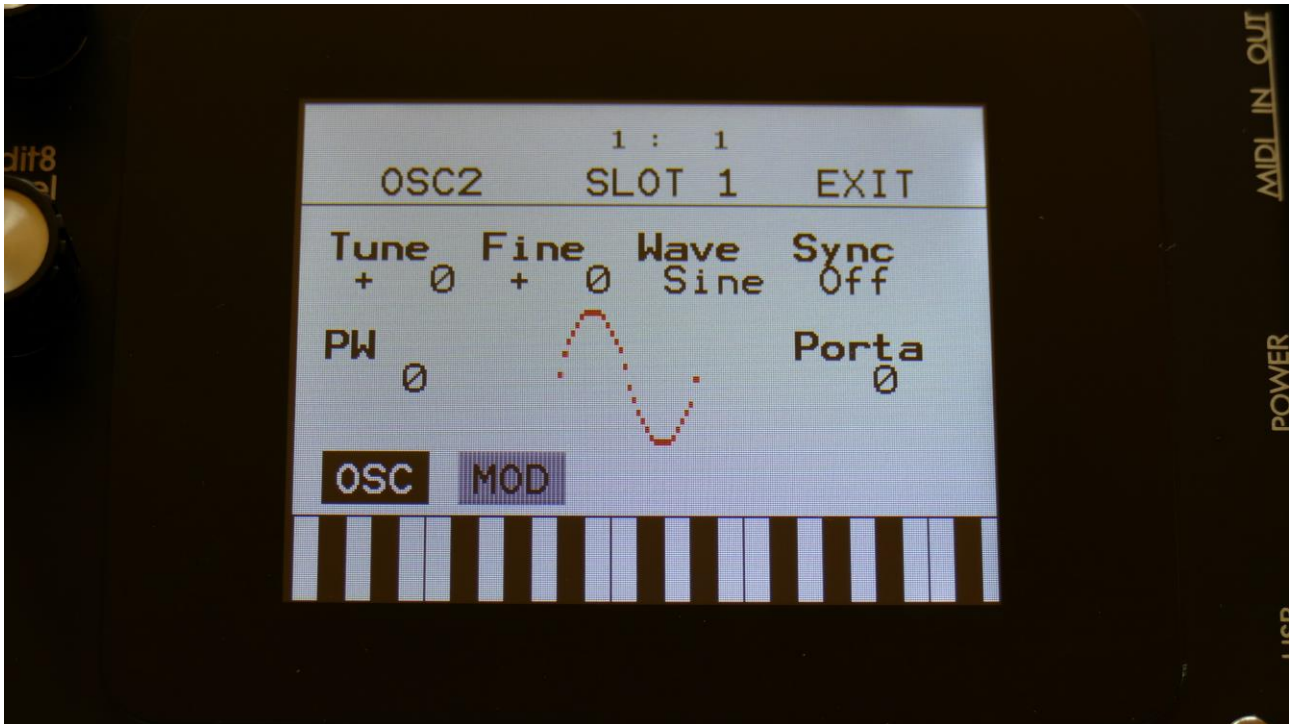
OSC2 and the Ring Modulator



Spazeboard 6 part 1 to 6 has an extra oscillator. Oscillator 2 generates a waveform that is switchable between sine, triangle, saw, square and noise waveforms. Pulse width are adjustable for sine, triangle, saw and square. Pitch and PW can be modulated. The pitch range of the oscillators are chromatically over the entire 10 octave MIDI keyboard range.

A Ring Modulator is placed on the output of Oscillator 2. The modulator for this can be selected on the Oscillator 2 modulation page.

Oscillator 2 parameters



In the middle of the oscillator page, the current waveform is shown.

By touching the buttons named “OSC” and “MOD”, it is possible to switch between the Oscillator2 page and the Oscillator2 Modulation pages.

Tune: Adjust the basic pitch in semitones. Range: -64 to + 63.

Fine: Fine tuning of the pitch. Range: -256 to +255.

Wave: Selects the oscillator 2 waveform. Choices are: Sine, triangle, saw, square, noise, audio input left/right, and Pitched noise (pnoi).

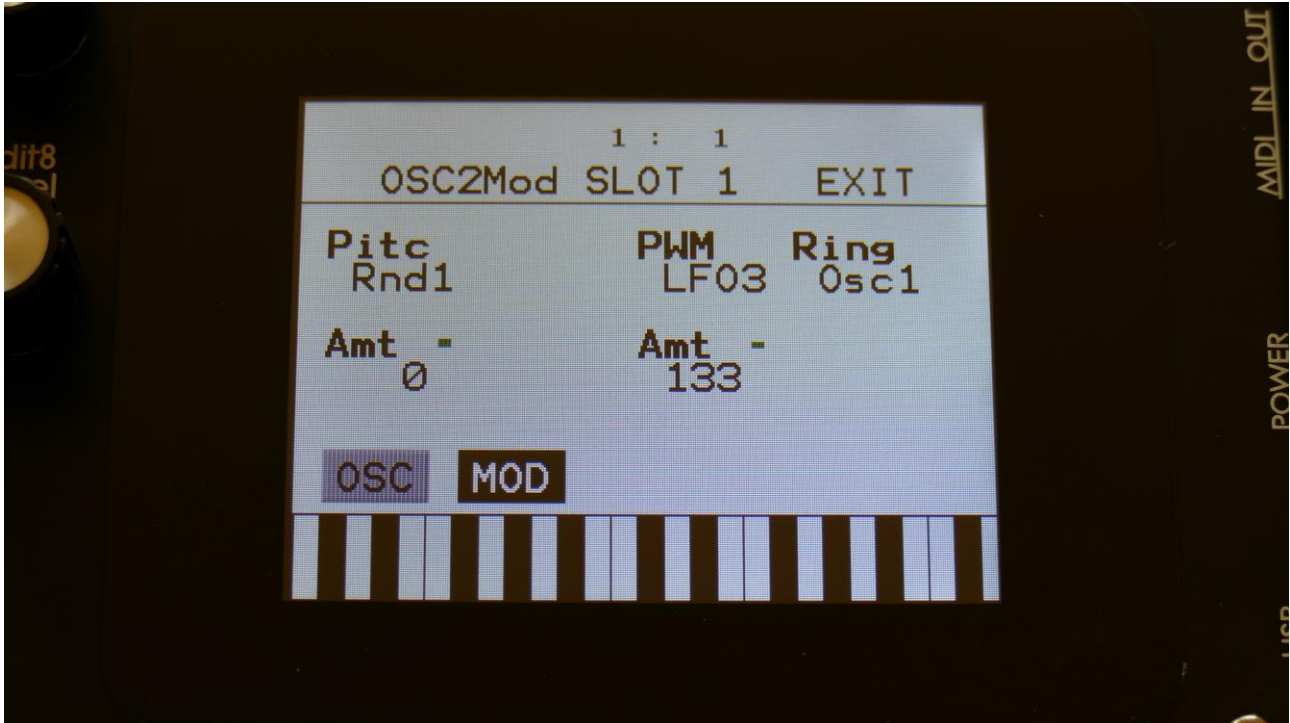
Sync: Oscillator Sync. When this is on, oscillator 2 will sync to oscillator 1.

PW: Adjusts the pulse width of the waveform. Pulse width can be adjusted for the sine, triangle, saw, square and pitched noise waveforms.

Porta: Portamento. The more this is turned up, the slower the oscillator pitch will slide from one note to another. Range: 0 to 511. Affects both osc1 and osc2.

Oscillator 2 modulation

Touch the MOD touch button, to enter the modulation page:



The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **The lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Pitc: Modulates the oscillator pitch.

PW: Modulates the pulse width of the waveform

Ring: Selects source 2 for the Ring Modulator. Source 1 is always Oscillator 2. Source 2 can be Oscillator 1 or any modulation source.

HPF and LPF –The Analog Filters

Spazeboard6 has 2 analog filters per Synth part, for part 1 to 6. A resonant highpass filter and a resonant lowpass filter.

The signal from Oscillator 1, Oscillator 2 and the Ring Modulator are mixed into the highpass filter. The signal from the highpass filter goes into the lowpass filter.

HPF cutoff and resonance, LPF cutoff and resonance and filter FM (FFM –Osc2 is the modulator) can be adjusted and modulated. It is also possible to adjust the mix of Osc1, Osc2 and the Ring Modulator, and to boost the filter.

The analog filters also has a G-Ray (Gotharman-Ray) digital/analog feedback circuit attached to them. This creates a kind of intermodulated feedback signal, and makes it possible to create sounds similar to FM plus new and never heard before sounds.

The parameters of the Analog Filters

From the synth parts main page, touch HPF/LPF to enter the analog filters. OR: Push and release the Func/Mute button, so that it lights up, and push step button 7/Filter 1 time.



Touch the VCF touch button, to enter this page.

The VU-meter at the right of the screen, shows the activity of the filter output.

HpCut: Adjusts the HPF cutoff frequency. Range: 0 to 511.

HpRes: Adjusts the resonance of the HPF. Range: 0 to 511.

LpCut: Adjusts the LPF cutoff frequency. Range: 0 to 511.

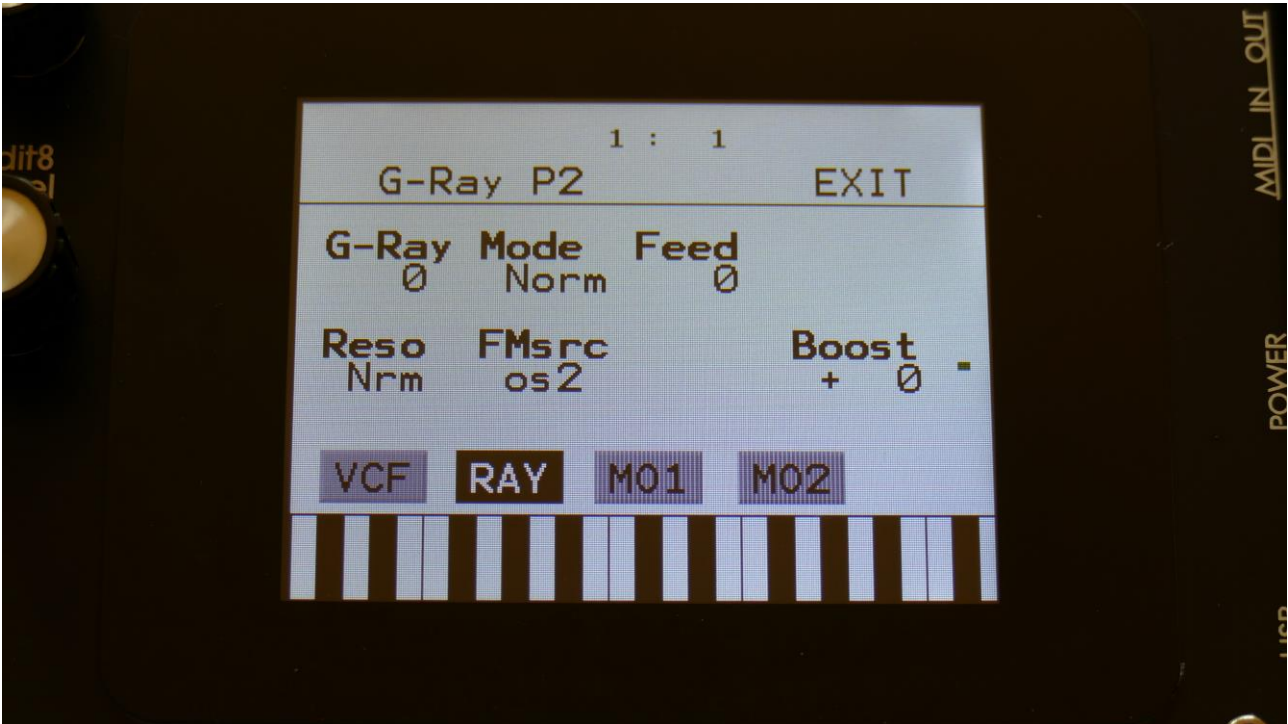
LpReso: Adjusts the resonance of the LPF. Range: 0 to 511.

Osc1, Osc2, Ring: Adjusts the levels of Oscillator 1, Oscillator 2 and the Ring Modulator, inputted to the analog filter.

The sources for the Ring Modulator can be set at the Oscillator 2 modulation page (see a little bit earlier in this manual).

FFM: Filter Frequency Modulation. Adjusts how much the analog filters should be audio frequency modulated by Oscillator 2.

Touch the RAY button, to enter the next filter page:



G-Ray: Adjusts the amount of g-RAY intermodulation. 0: no g-RAY, 3: max g-RAY. Range: 0 to 3.

Mode: G-Ray mode.

-**Norm:** Normal 1:1 feedback.

-**Neg:** 1:1 feedback with the signal inverted (a 180 degree phase shift)

-**Ultr:** Boosted feedback.

-**Uneg:** Boosted feedback with the signal inverted (a 180 degree phase shift)

Feed: G-Ray feedback level. Range: 0 to 511.

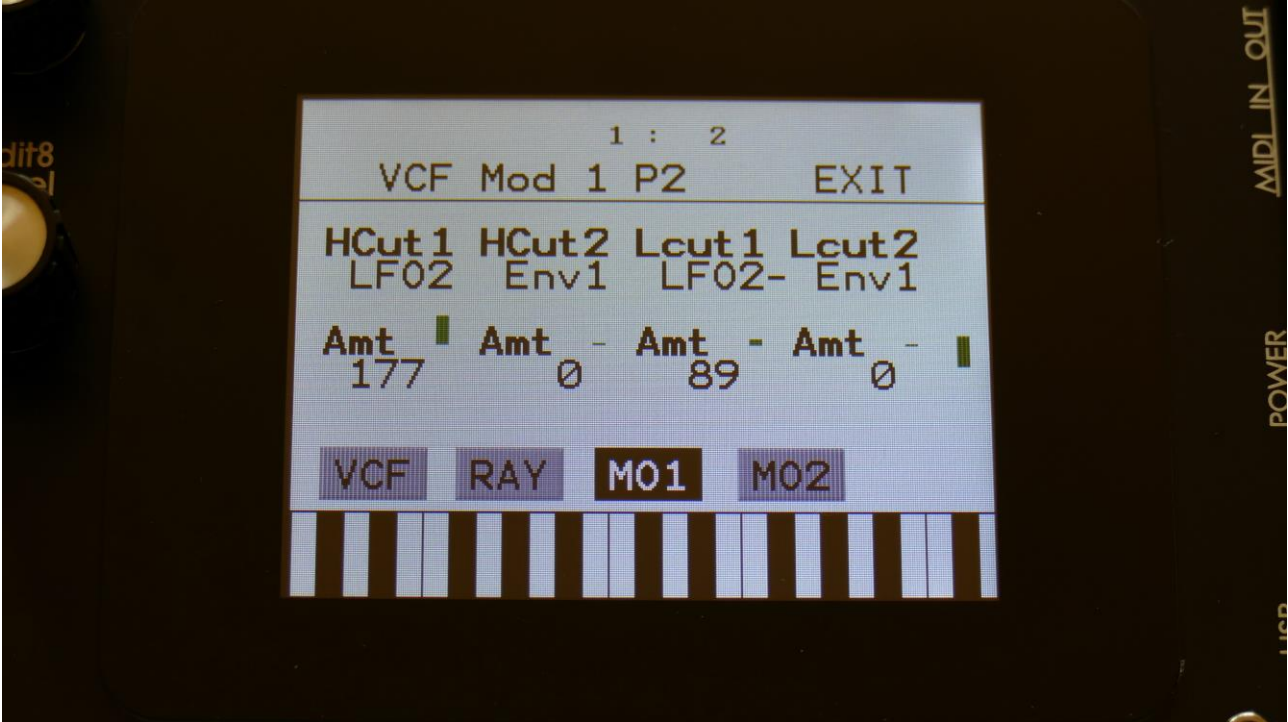
Reso: Selects whether the maximum resonance for the analog filters of this part should be normal (Nrm) or maximum (Max). At the normal setting, the resonance setting is more precise, making it ideal for presets that should be shared, or for sounds that should be copied between parts. The Max setting adds more resonance to the sound, but it is also more unprecise, so if you share the sound between multiple Spazeboard6's or copies the sound to another part, you might have to re-adjust the resonance, to get the exact same sound.

FMsrc: Selects whether the FM source, for both the analog filters and Oscillator1, should be the audio output of Oscillator2 (os2), or the audio output of the digital filter (dgf).

Boost: If necessary, it is possible to boost or lower the output level of the analog filters. The range of this is -128 to +383, with +0 being the neutral and initialized setting.

Analog Filters Modulation 1

Touch the MO1 button, to enter this first modulation page:



The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **The lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

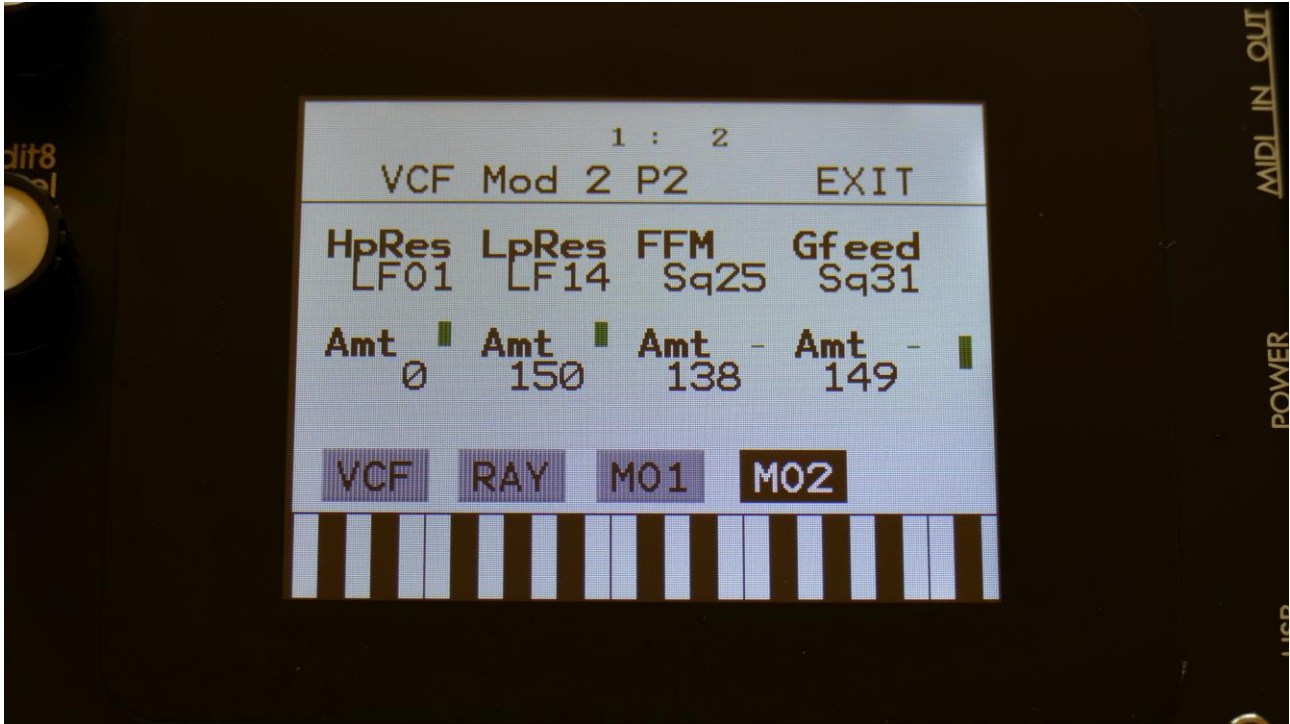
The parameters on this page:

HCut1 and HCut2: Modulates the HPF Cutoff Frequency.

LCut1 and LCut2: Modulates the HPF Cutoff Frequency.

Analog Filters Modulation 2

Touch the MO2, to enter this second filter modulation page:



The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **The lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

HpRes: Modulates the HPF Resonance.

LpRes: Modulates the LPF Resonance.

FFM: Modulates the amount of Filter FM.

Gfeed: Modulates the amount of G-Ray feedback.

The Digital Filter (DGF)

Spazeboard6 part 1 to 8 each has a resonant digital filter. 16 different filter types can be selected. They have various lowpass, highpass and bandpass modes, with different characteristics, but also special modes like Lo-Fi, destruktion and Fat. On part 1 to 6, they can be placed before, after or in parallel with the 2 analog filters. It is also possible to route Oscillator 1 to the analog filters, and Oscillator 2 to the digital filter, in "Split" mode.

On part 7 and 8, oscillator1 is connected to the digital filter input.

When different filter types are selected for Morph Layer A and B, LD3 will morph between the filter types.

For extra filtering, the same filters are available as insert effects. The filters in the effect processors does though not morph between different filter types.

List of Digital Filter Types:

LPF1: Lowpass filter with a rather weak character. Resonance does not self-oscillate.

LPF2: Lowpass filter that are a bit sharper than LPF1. Resonance does not self-oscillate.

LPF3: Sharp Lowpass filter with self-oscillating resonance.

LPF4: Very sharp Lowpass filter with self-oscillating resonance.

BPF1: Bandpass filter with a rather weak character. Resonance does not self-oscillate.

BPF2: Bandpass filter with focus on the bass area. Distorts at higher input levels. Resonance does not self-oscillate.

BPF3: Sharp Bandpass filter with self-oscillating resonance.

HPF1: Sharp Highpass filter with self-oscillating resonance.

HPF2: High gained Highpass filter with self-oscillating resonance. Distorts at higher input levels.

HPF3: Sharp Highpass filter with self-oscillating resonance. A bit weaker than HPF1.

Dstr: Destruktion. A rather defective filter. Self-oscillates at some points, distorts at others.

FAT1: Slightly distorting lowpass filter, with a sharp response and self-oscillating resonance.

FAT2: A slightly weaker version of FAT1.

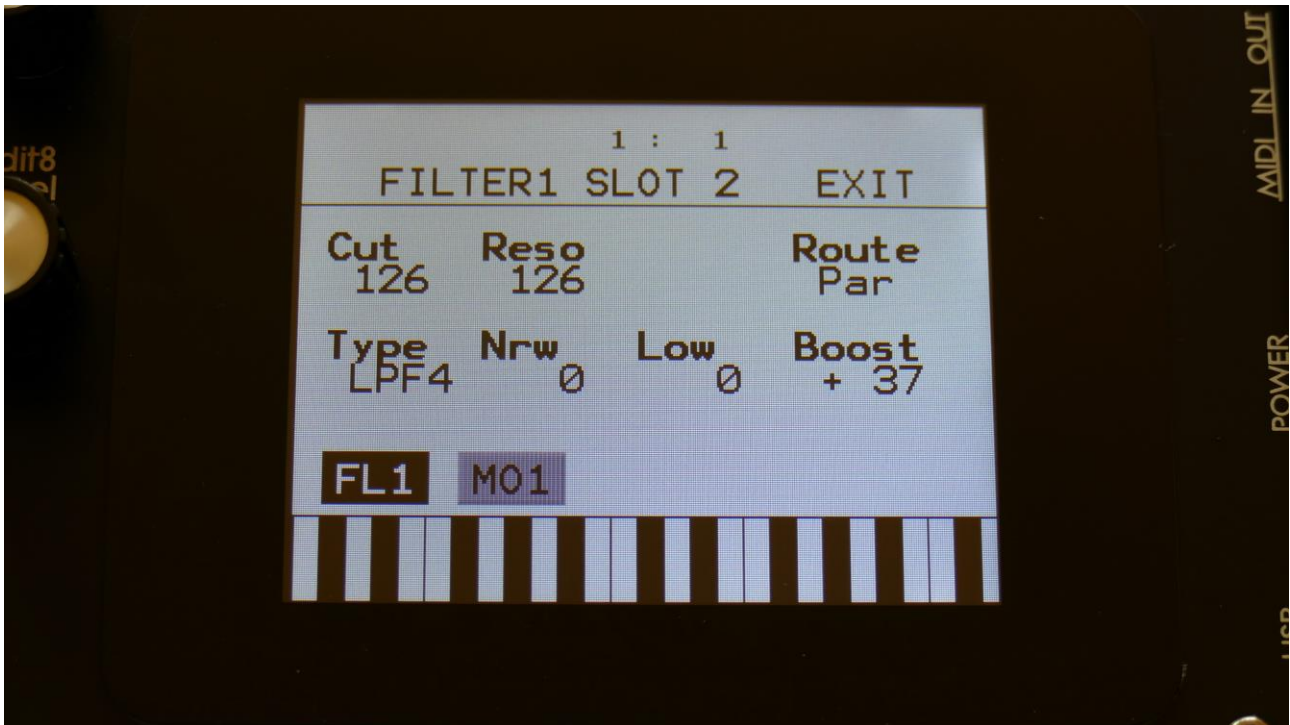
LoFi: A very distorting and unpredictable lowpass filter.

LPF5: A lowpass filter with a very soft character. Resonance does not self-oscillate.

BPF4: A bandpass filter with a very soft character and self-oscillating resonance.

Digital Filters Parameters

From the synth parts main page, touch DGF to enter the digital filters. OR: Push and release the Func/Mute button, so that it lights up, and push step button 7/Filter 2 times.



Touch the FL1 touch button, to enter the first DGF page.

Cut: 0 to 511. Sets the cutoff frequency of the filter.

Reso: 0 to 511. Sets the amount of resonance applied to the filter.

Route: The routing of the digital filter. Can be set to:

Par: The digital filter is positioned in parallel to the analog filters.

Split: The digital filter is again positioned in parallel to the analog filters, but Osc1 only goes to the analog filters, and Osc2 only goes to the digital filter.

Pre: The analog and digital filters are connected in serial, with the digital filter first.

Post: The analog and digital filters are connected in serial, with the analog filter first.

On Part 7 and 8, Route is replaced by Inp: 0 to 511. Sets the audio signal input level to the filter. Different characteristics can be obtained by adjusting this.

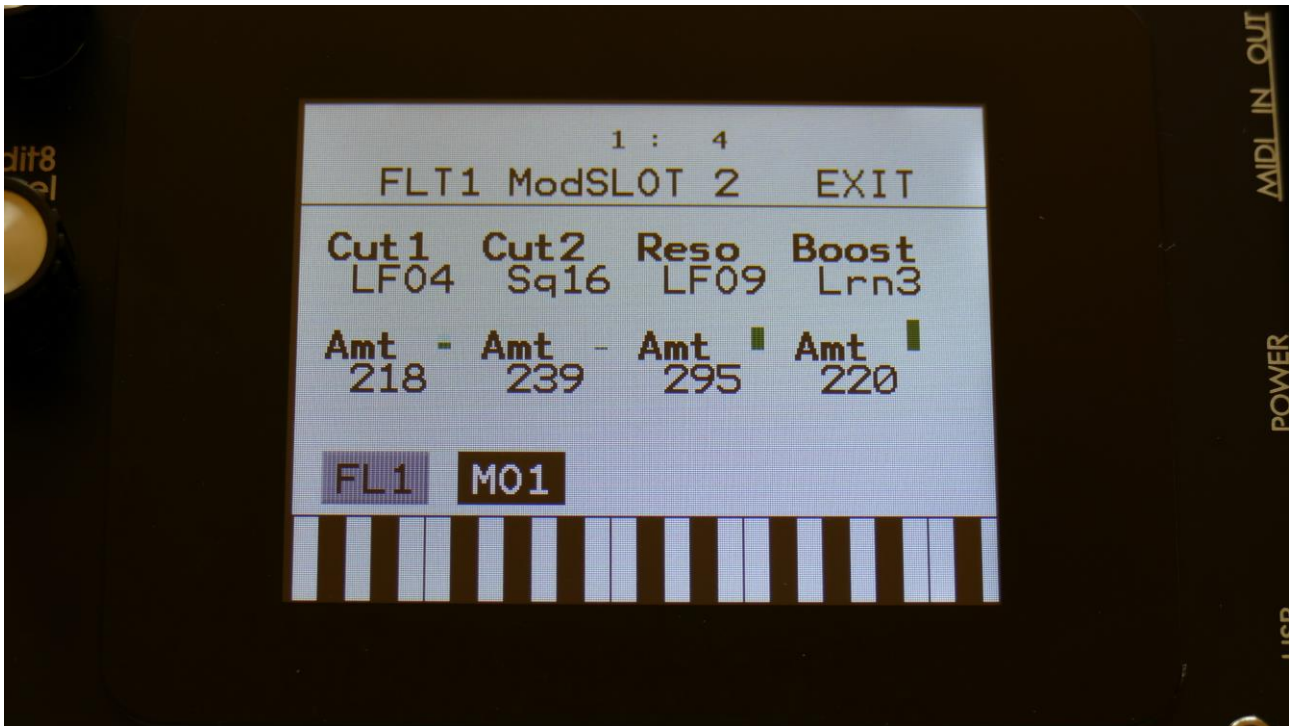
Type: Sets the filter type. See the list of filter types.

Nrw: 0 to 511. Turning this up, will make the frequency response of the filter more and more narrow.

Low: 0 to 511. Sets the lowest frequency offset point of the filter. Turning this up, will in many filter types, make the bass bottom more present or distorted.

Boost: Boost/level adjustment of the digital filter. The range of this is -128 to +383, with +0 being the neutral setting.

Digital Filter Modulation



Touch the MO1 touch button, to enter the Digital Filter Modulation page.

The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

Cut1, Cut2: Will modulate the filter cutoff frequency.

Reso: Will modulate the amount of resonance applied to the filter.

Boost: Will modulate the digital filter output level/gain.

VCA

The VCA is the last stage of a synth part, before the audio is sent to insert effects in output mode, the output effects, or the stereo outputs. The audio output from the analog and digital filters, goes into the VCA.

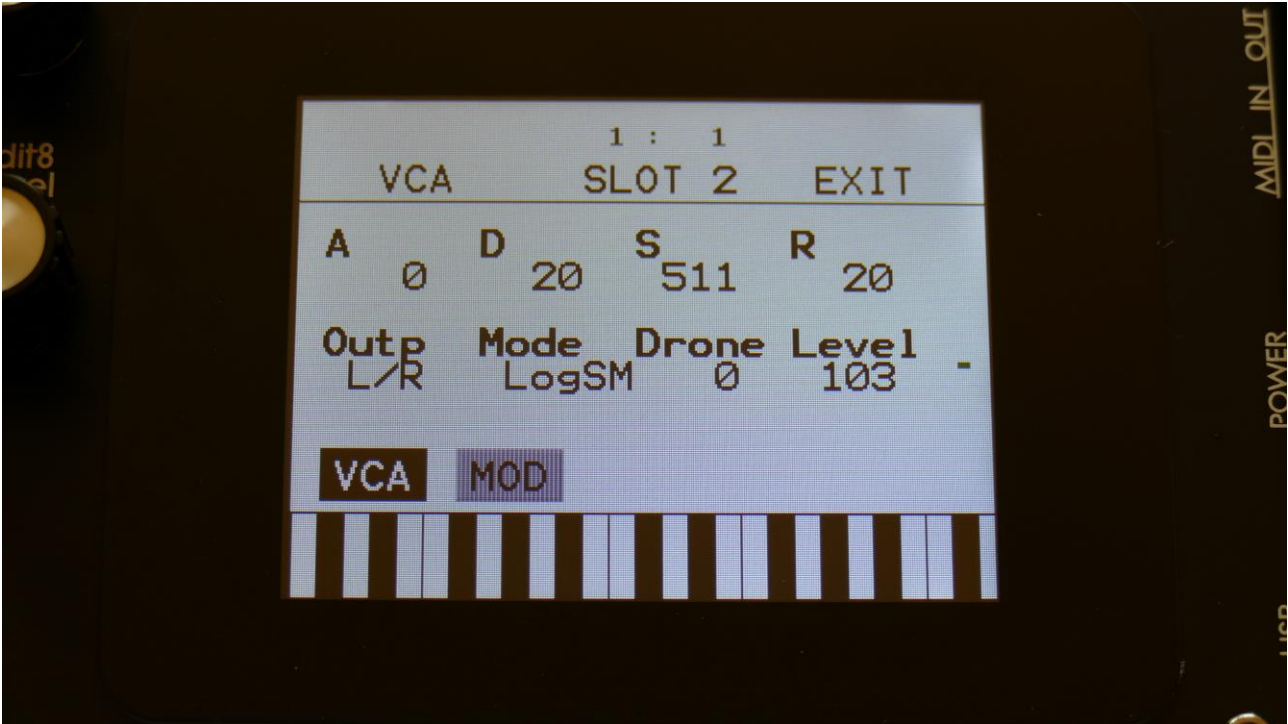
The VCA's can behave in 4 different modes. Linear, logarithmic, smooth linear and smooth logarithmic. The smooth VCA modes behaves like the VCA's found in most other synthesizers.

The audio output of the VCA can be level modulated and panned. Pan can also be modulated. The output of the VCA can be sent to both output effects at the same time. When set up like this, the pan control/modulation, is panning the audio signal between the 2 effects.

An ADSR envelope are attached to the VCA. The VCA envelopes can either be in linear or logarithmic mode. A Drone parameter are available, for opening the VCA without the envelope needing to be triggered. VCA output level are modulated by the attached ADSR envelope. Attack and release can be modulated by any modulation source.

VCA Parameters

From the synth parts main page, touch VCA to enter the VCA pages. OR: Push and release the Func/Mute button, so that it lights up, and push step button 8/VCA 1 time.



Touch the VCA touch button, to enter the VCA parameters page.

The VU-meter at the right of the screen, shows the VCA output activity.

A: 0 to 511. VCA envelope attack time. The time it will take the VCA envelope to rise from zero to its maximum value, when a note event is received and held down.

D: 0 to 511. VCA envelope decay time. When the VCA envelope has reached its maximum value, in the time set by the attack parameter, it will decay, until it reaches the sustain level, and stay there, as long as the note that triggered it is held.

S: 0 to 511. VCA envelope sustain level. Explained under the "D" parameter.

R: 0 to 511. VCA envelope release time. The time it will take the VCA envelope to decay from the value it is at, when a note off event are received, to zero.

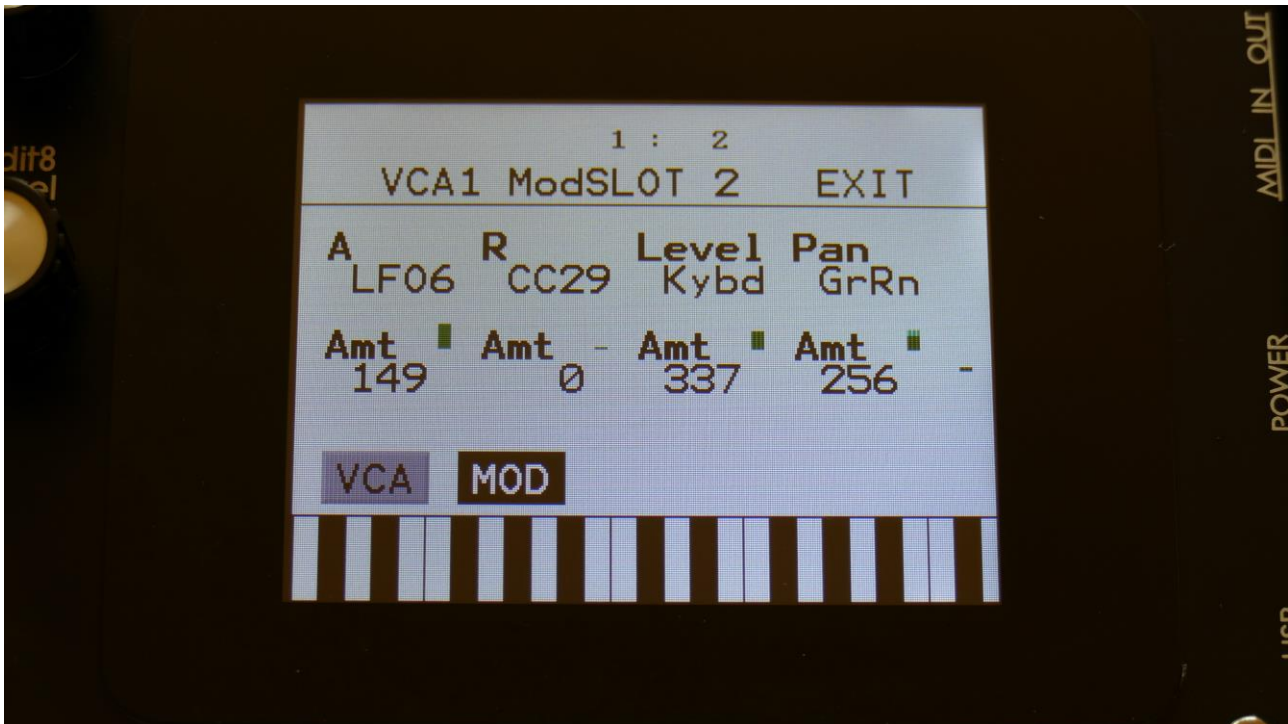
Outp: Sends the audio output of the VCA to the left and right audio outputs (L/R), or to the output effects (EFX).

Mode: Selects if the VCA behaviour should be linear (**Lin**), logarithmic (**Log**), smooth linear (**LinSM**) or smooth logarithmic (**LogSM**). The logarithmic curve gives the sound a softer and less “clicky” attack. The smooth modes produces no clicking at all.

Drone: 0 to 511. VCA envelope drone offset level. When this is turned up, the VCA envelope will never reach an output value, lower than what this is adjusted to –It will release to this adjusted value, instead of zero. Use this to keep the output of a synth part open for drone sounds.

Level: The VCA output level.

VCA Modulation



Touch the MOD touch button, to enter the VCA modulation page. OR: Push and release the Func/Mute button, so that it lights up, and push step button 8/VCA 2 times.

The VU-meter at the right of the screen, shows the VCA output activity. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

A: Modulates the VCA Envelope attack time.

R: Modulates the VCA Envelope release time.

Level: Modulates the output level of the VCA.

Pan: Modulates the output panning of the VCA. When source is set to manual ("Man"), the amount parameter manually adjusts the output panning.

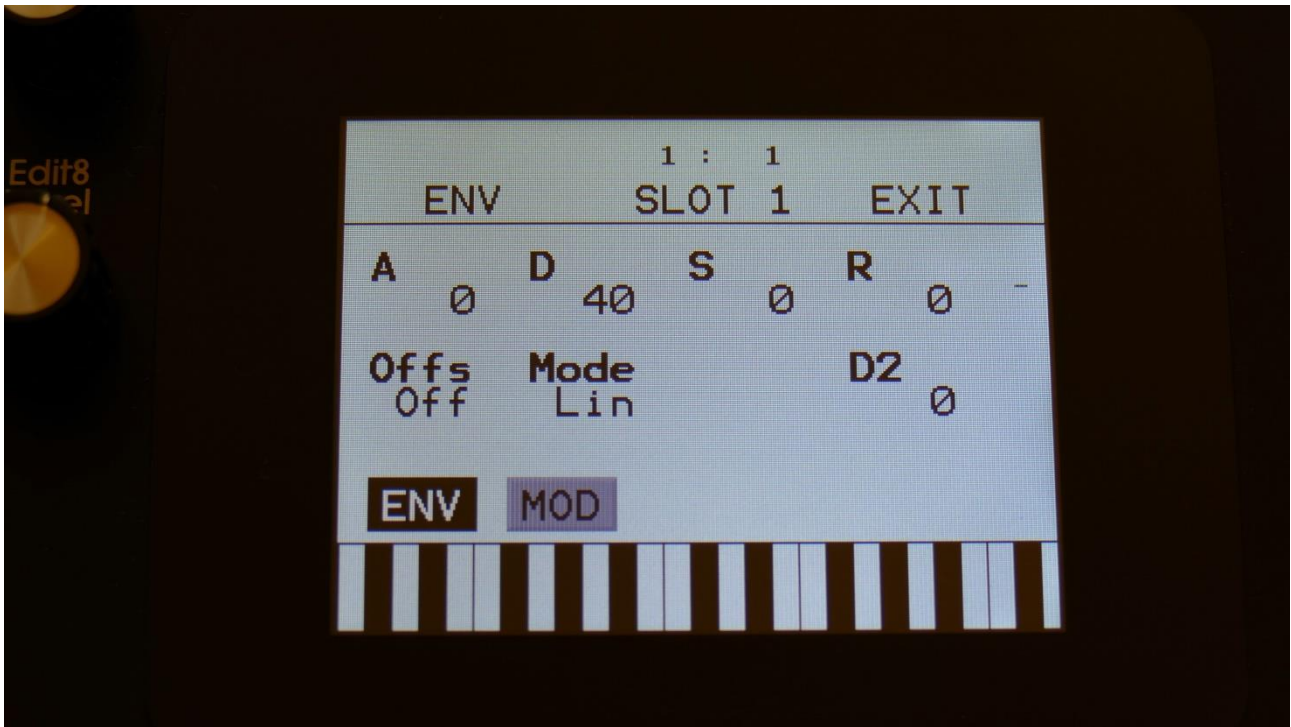
Envelope 1 and 2

Each Synth part of Spazeboard6 has 2 modulation envelopes. One ADSR type, and one decay only envelope. Envelope 1, the ADSR envelope can have linear or logarithmic characteristics and it also has an offset control.

Envelope 2, the decay envelope, is always linear.

Envelope 1 and 2 Parameters

From the synth parts main page, touch ENV to enter the envelope pages. OR: Push and release the Func/Mute button, so that it lights up, and push step button 9/Env 1 time.



Touch the ENV touch button, to enter the envelope parameters page.

The VU-meter at the right of the screen, shows the Envelope output.

A: 0 to 511. Envelope 1 attack time. The time it will take the envelope to rise from zero to its maximum value, when a note event is received and held down.

D: 0 to 511. Envelope 1 decay time. When the envelope has reached its maximum value, in the time set by the attack parameter, it will decay, until it reaches the sustain level, and stay there, as long as the note that triggered it is held.

S: 0 to 511. Envelope 1 sustain level. Explained under the "Dec" parameter.

R: 0 to 511. Envelope 1 release time. The time it will take the envelope to decay from the value it is at, when a note off event are received, to zero.

Offs: Offset:

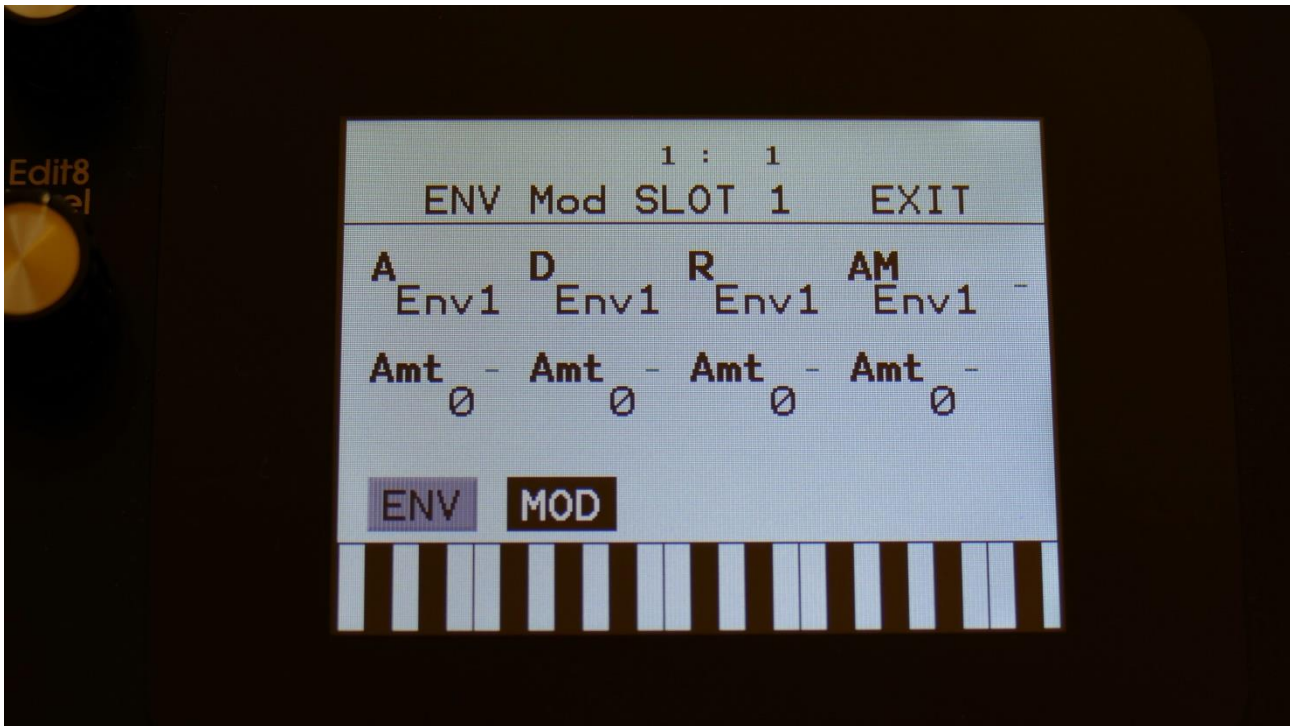
-Off: The envelope will work around the zero point, and apply both negative and positive modulation to the parameters affected by it.

-On: Positive only, offset added. The envelope will only work above the zero point, and will only add to the values of the parameters affected by it.

Mode: Selects if the envelope curve should be linear (**Lin**), or logarithmic (**Log**). The logarithmic curve gives the envelope a softer and less “clicky” attack.

D2: 0 to 511. Envelope 2 decay time. The time it will take the decay envelope to decay, after it has been triggered.

Envelope Modulation



Touch the MOD touch button, to enter the envelope modulation page. OR: Push and release the Func/Mute button, so that it lights up, and push step button 9/Env 2 times.

The VU-meter at the right of the screen, shows the Envelope output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

A: Modulates the Envelope 1 attack time.

D: Modulates the Envelope 1 decay time.

R: Modulates the Envelope 1 release time.

AM: Modulates the Envelope 1 output amount.

LFO1 to 16

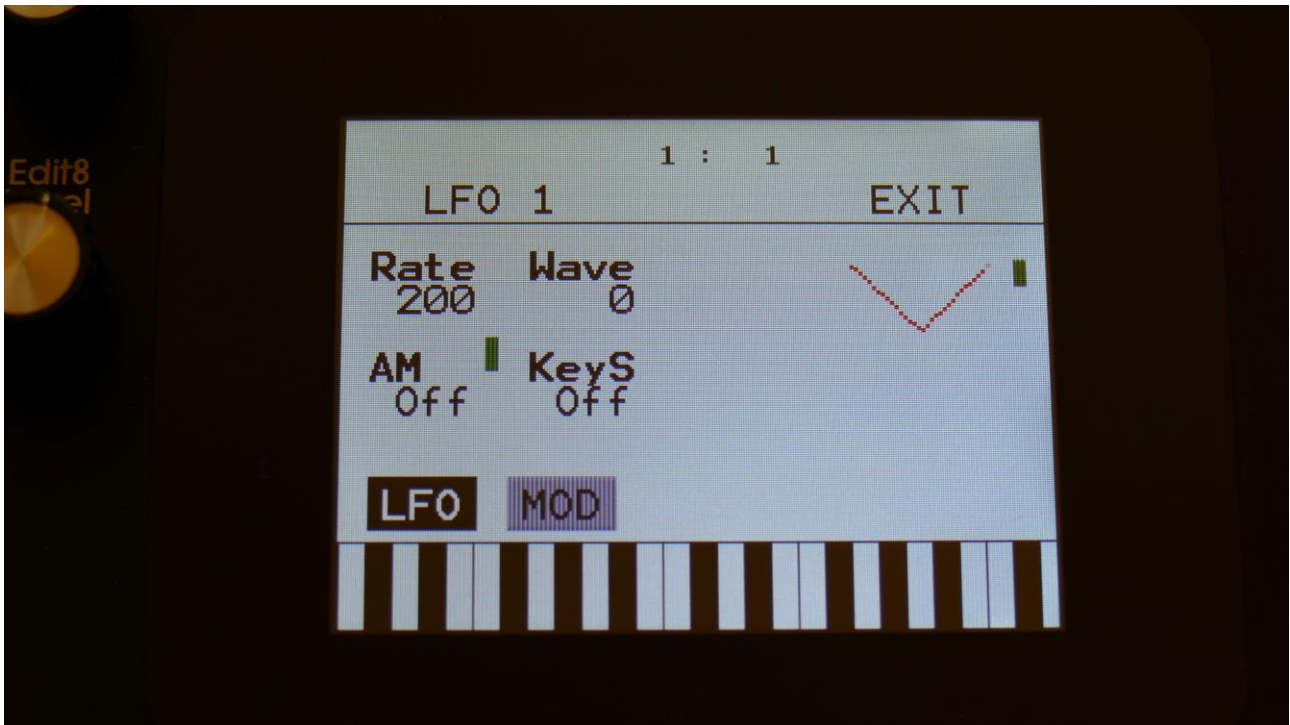
LD3 has 16 LFO's, that are global, so each LFO can modulate any of the 16 Synth parts.

The LFO's has continuously variable waveform and rate parameters. Both of these parameters can be modulated. The output level of the LFO's can also be modulated, and they can be key-synced.

One LFO are shown at a time on the LFO pages. LFO 1 to 16 are selected in the same way as synth parts are, by pushing and holding the Steps/Part button, while pressing a step button. The LFO's are though selected separately from the parts. So LFO 5 can for instance be selected, while part 3 is selected.

LFO parameters

From the synth part main page, touch the LFO block, to enter the LFO pages. OR: Push and release the Func/Mute button, so that it lights up, and push step button 10/LFO 1 time.



Touch the LFO touch button, to select this page.

At the right of this page, the current LFO waveform is shown, together with a small VU-meter, showing the LFO output activity.

Rate: 0 to 511. LFO rate. Sets the speed of the LFO.

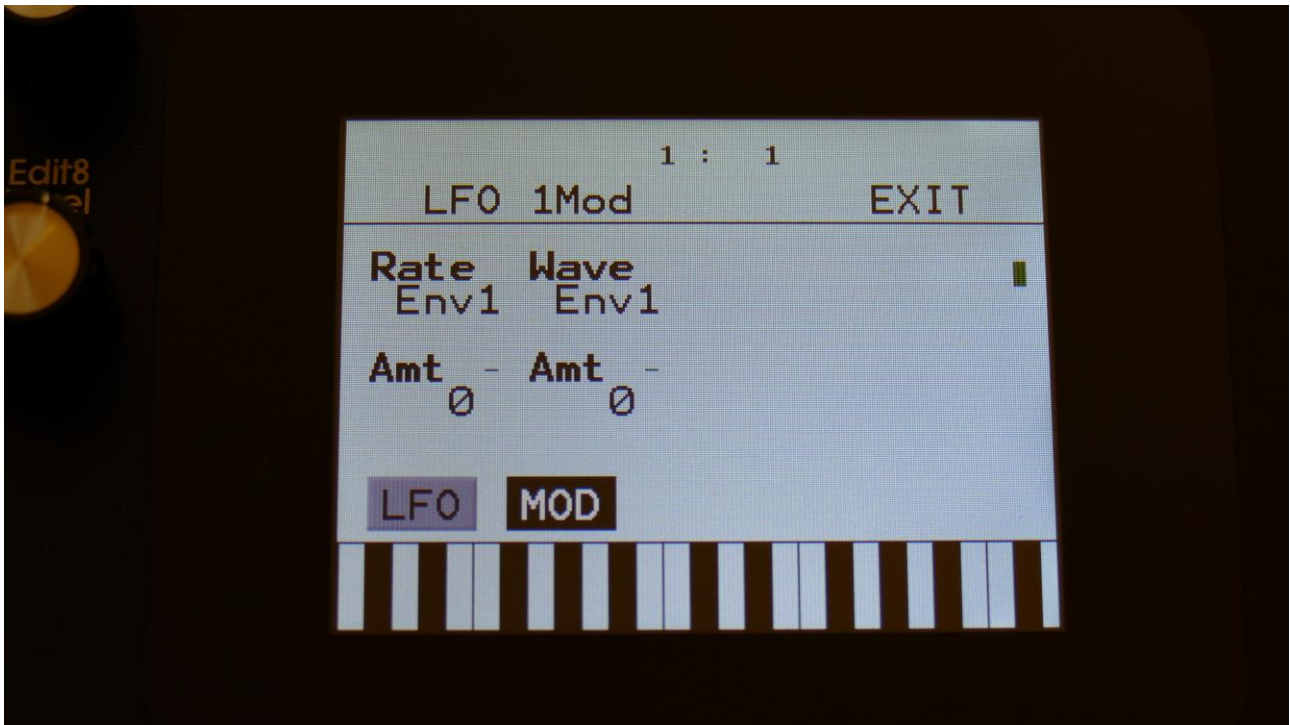
Wave: 0 to 511. LFO output waveform. Morphs between triangle, sawtooth, square, pulse and FM waveforms. The FM waveforms are high frequency waves, that can be used for FM synthesis.

AM: Selects the modulation source, that will control the LFO output level. If "Off" is selected, no output level control will be applied to the LFO. Any modulation source can be selected. Please see list of modulation sources in the beginning of this section.

KeyS: Off, 1 to 16: LFO key sync.

Off: The LFO is free running.

1 to 16: Every time this Synth part is triggered, the LFO will reset to its initial value, and start over.



Touch the MOD touch button, to select this page. OR: Push and release the Func/Mute button, so that it lights up, and push step button 10/LFO 2 times.

The VU-meter at the right of the screen, shows the LFO activity. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Rate: Modulates the LFO rate.

Wave: Modulates the LFO waveshape.

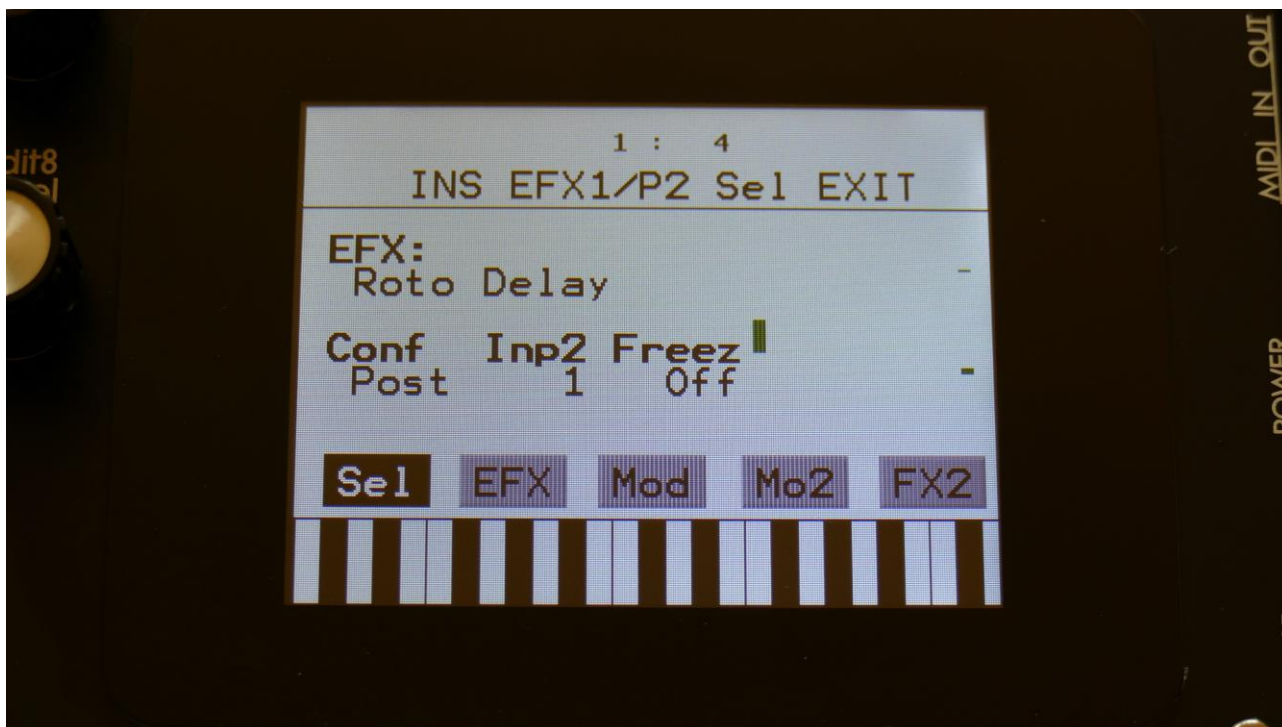
Effects Processors

Spazeboard6 has 12 insert effect processors (2 for each part of part 1 to 6) and 2 output effect processors, 14 effect processors in total. Each insert effect can be placed either before or after the VCA of part 1 to 6. Part 1 to 8 can be sent to the output effects. The output effects outputs to the main L and R outputs.

Insert Effects 1 and 2

Selecting the Insert effect processors

From the synth parts 1 to 6 main page, touch EFX1 or EFX2 to enter an effect page. OR: Push and release the Func/Mute button, so that it lights up, and push step button 11/EFX 1 time.



To switch between Insert effect 1 and 2 of the part, touch the FX2/FX1 touch button in the lower right corner of the screen.

All effects has a bypass switch, Freeze modulation, and a mix parameter.

List of Insert Effects

Chorus – Gotharman's special chorus with an added Deep parameter, that adds space to the chorus.

Distortion – 4 types: Valve, Sine, Fuzz, Xdis.

Bit Crush – Lowers the sample rate and the bit resolution of the sound, to obtain lo-fi effects.

Pitch Shifter – Shift the pitch of the sound up to 4 octaves up or down, without changing the time resolution or “tempo” of the sound. Adjustable sense.

Resonator – Simulates the resonances that comes, if a sound goes through a small box. LD3's resonators are synthetic, with more focus on making sounds, than on simulating actual boxes.

Stretcher – Tries to time stretch the input signal, while at the same time keeping up with it. Impossible? -Yes, indeed :-)

FM – Adds self-FM to the input signal in +/- 1 octave, +/- 2 octaves or +/- 4 octaves ranges.

Glitch Shifter – Imperfect pitch shifter.

Glitch Shifter 2 – Imperfect pitch shifter with a slightly different sound than the first one.

Glitch Shifter 3 – Imperfect pitch shifter with a slightly different sound than the first two.

Pitch Shaper – 1 input version of Gotharman's special Pitch Shaper, that forces an audio signal to play back at a specific pitch, determined by an adjustable frequency.

FAT - Adds up to 3 layers of the sound to itself, and it is possible to adjust the phase of these, and to select whether the effect sound should be boosted or just layered.

Filter - This is exactly the same filters, as can be found in the synth parts. They were added to the insert effects section, in order to make it possible to filter external signals, to build an LD2 style filterbank, using multiples of these, and because sometimes it is handy to filter some parts of the other effects or the analog filters out.

Filters 2 - A different set of digital filters, that is emphasized in the bottom bass area, and has a pretty uncontrollable resonance.

Compressor - Damps audio signals that are above a settable level, in order to make the signal less dynamic, and easier to record. This is not the same effect as the compressor in LD2. The LD2 compressor has been renamed “Expandor”.

Expandor - Gains audio signals that are above a settable level, in order to add more kick to sounds. This is the effect, that was named “Compressor” on the LD2.

PitchShaper 2 - This is a dual input PitchShaper. The pitch of the signal applied to input 1, is pitch matched to the signal applied to input 2. When changing the pitch of the pitch matched signal, the waveform on the effect output will change, still matched to the pitch of input 2.

It is also possible for the signal applied to input 1, to affect the output pitch, by turning the Vari parameter up.

The audio bus that this effect is placed on, applies the signal to be re-pitched. The audio bus numbered above the bus, that the effect is placed on, applies the signal that the first signal should be pitch matched to.

Delay 1 – Delay with time and feedback controls, plus Gotharman's Deep and Xfade controls. Deep adds space to the delay. The Xfade control on this delay, creates valleys between the delay taps. The more it is turned up, the more time the valleys takes up.

Roto Delay – This is a 2 tap delay, that is constantly crossfading between the 2 taps. When the Xfade control is turned down, the crossfading is rough, the more it is turned up, the more smooth the crossfading gets. Other controls are the same as the first delay.

Bright Delay – First delay, but with a brighter sound, created by a resonator.

Bright Roto Delay – Roto delay, but with a brighter sound, created by a resonator.

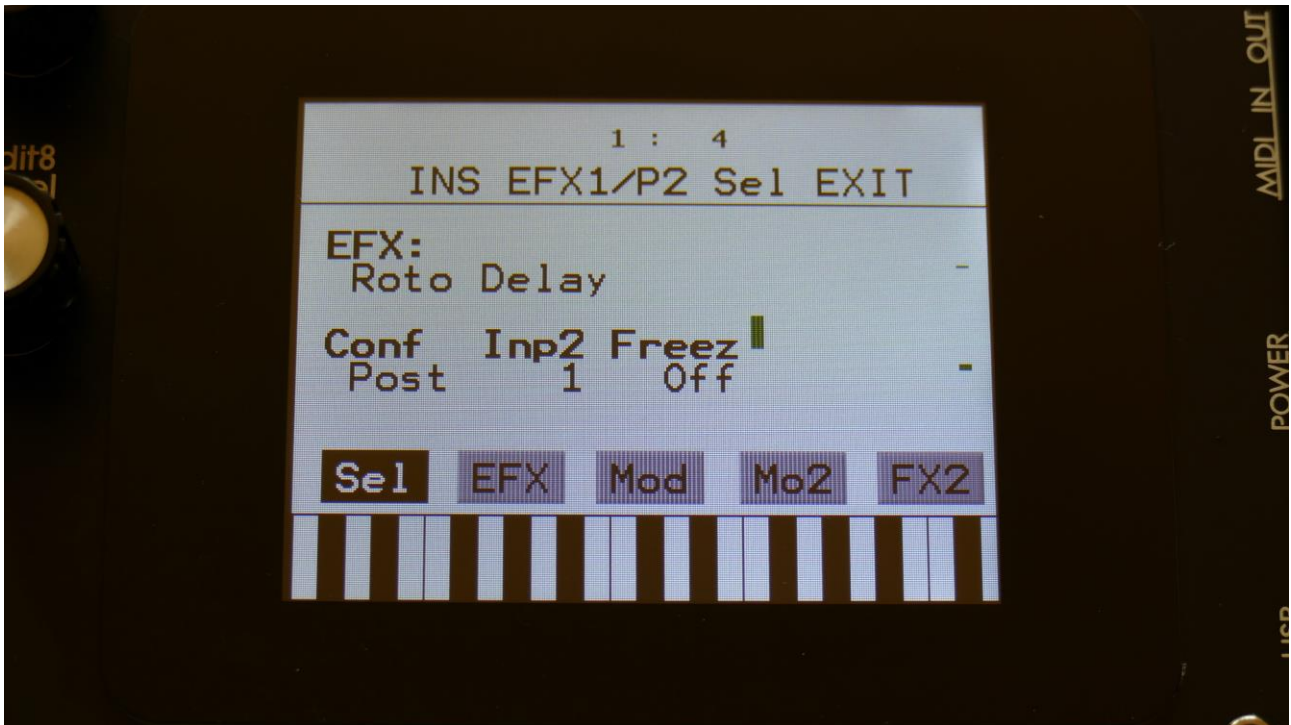
Variator – Creates new variations of the input signal. Both pitch and rhythmic variations.

Time Stretch - This effect is able to change the length of the input signal, almost without changing the pitch. Old fashioned time stretch done in realtime, that can be applied even to live input signals! It is also possible to change the pitch if desired, and it is possible to trigger this effect, for bigger precision.

Sample Pitch - Changes the pitch of a sampling, without changing the length, using granular technology. It is possible to set the size of each pitch grain, to generate great granular effects. This effect has to be triggered from any of the parts, in order to function properly. It will work on external signals, but it still needs to be triggered by a part.

Insert effects Select page

From the synth parts main page, touch EFX1 or EFX2 to enter the effect pages. OR: Push and release the Func/Mute button, so that it lights up, and push step button 11/EFX 1 time.



Touch the SEL touch button, to enter the effect select page. Touch FX2/FX1 to jump between insert effect 1 and 2.

The 2 VU-meters at the right of the screen, shows the Effect input and output activity.

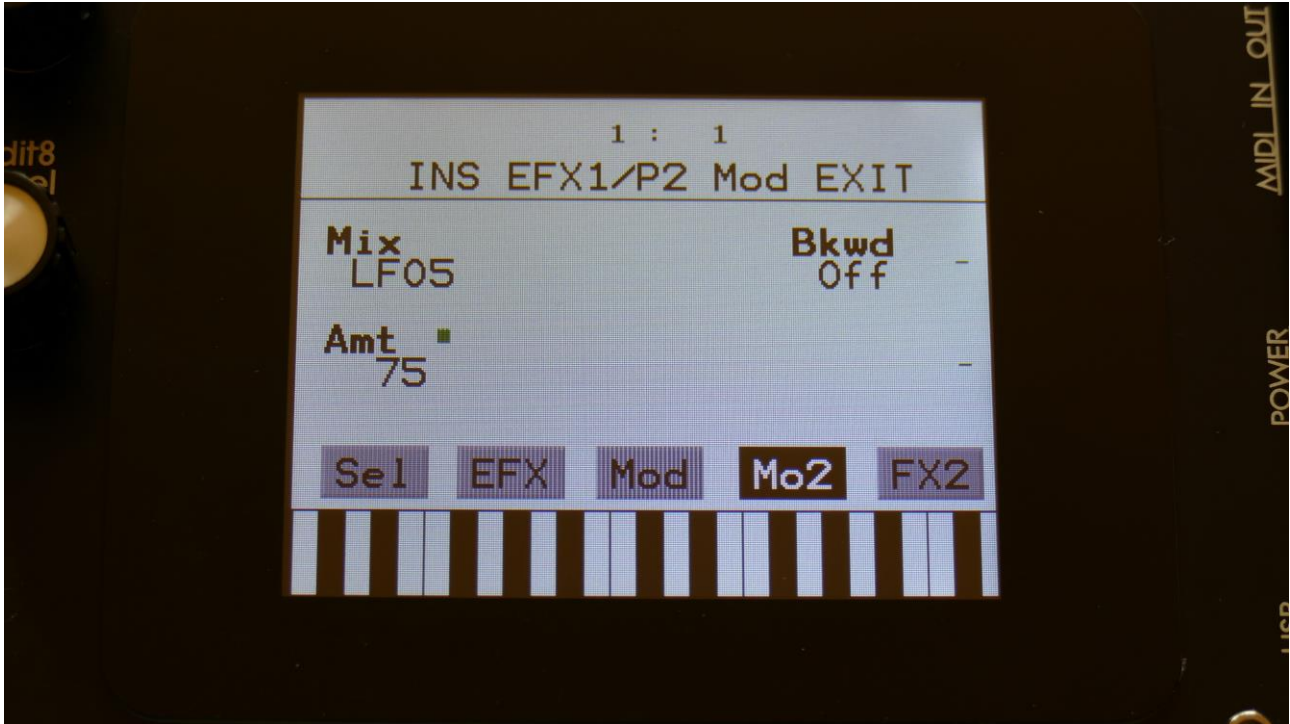
Conf: Places the Insert effect pre or post the VCA.

Inp2: 1 to 8. Select from which part the input 2 on the PitchShaper 2 effect should be taken.

Freez: Freeze modulation. Can be set to off, Freeze button (Buton) or to any modulation source. When set to any other value than off, the effect will freeze, when the modulation source are in its upper range. When an effect freezes, it no longer samples its audio input signal, but only outputs whatever it already holds in its audio buffer.

Insert Effects Parameters

Mix Modulation and Backwards Delay



The Effect dry/wet mix can be modulated on all effects.

Touch the MO2 touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

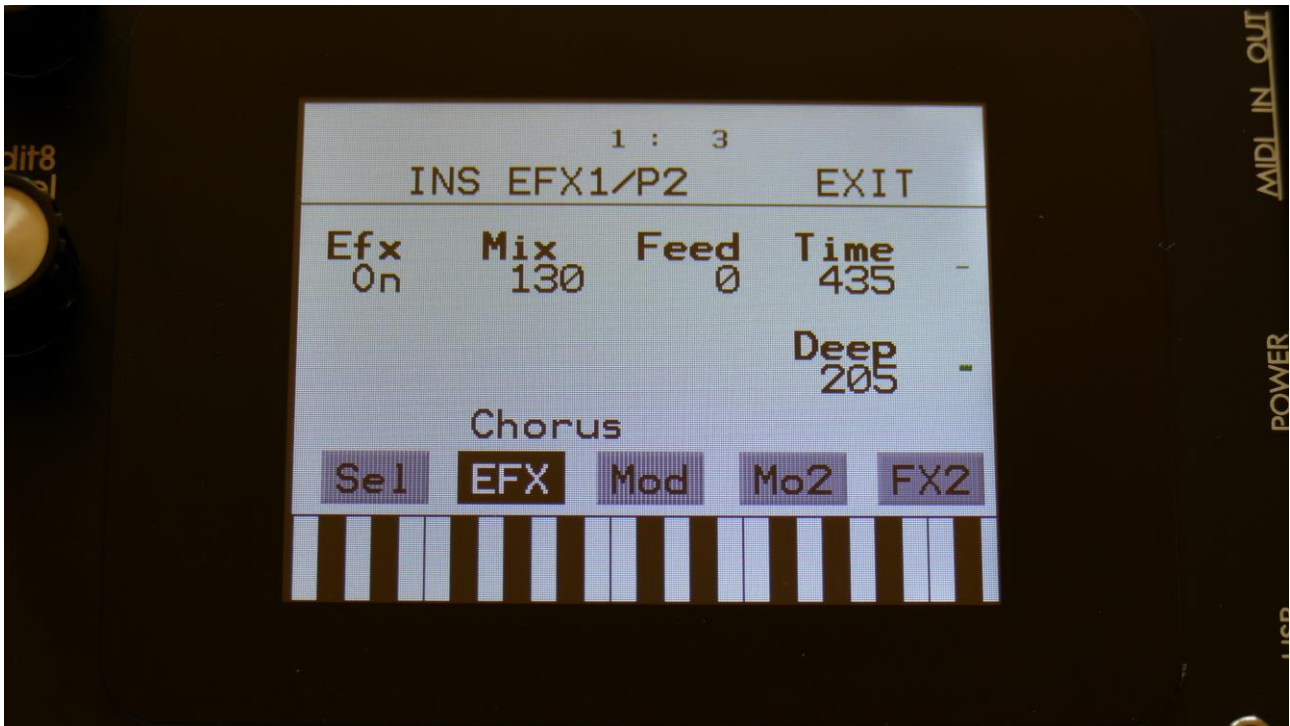
The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Mix: Modulates the effect dry/wet mix.

Bkwd: This is only visible, when a delay effect is selected. Turning this on, will make the delay effect play back backwards.

Chorus



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

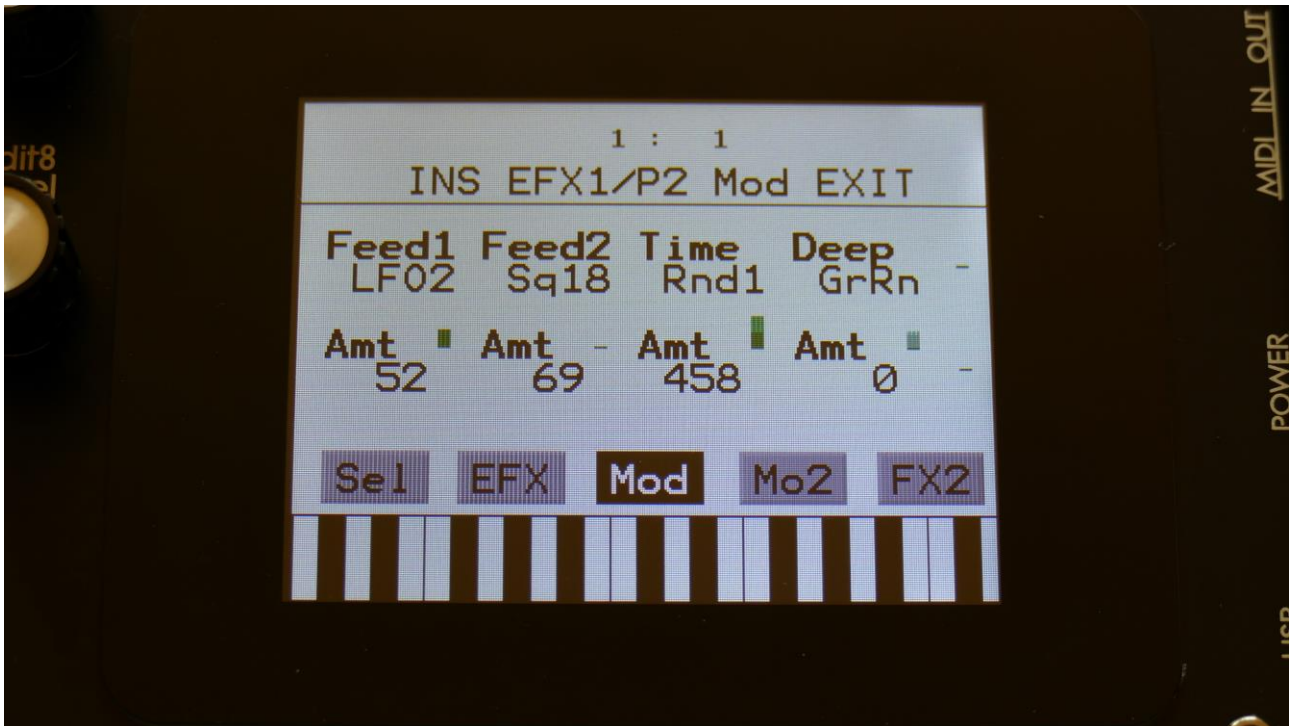
Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Feed: Chorus feedback amount.

Time: Chorus Time. This should be modulated by an LFO, to get the traditional chorus effect.

Deep: Adjusts how deep the chorus box should be. A Gotharman special.

Chorus Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

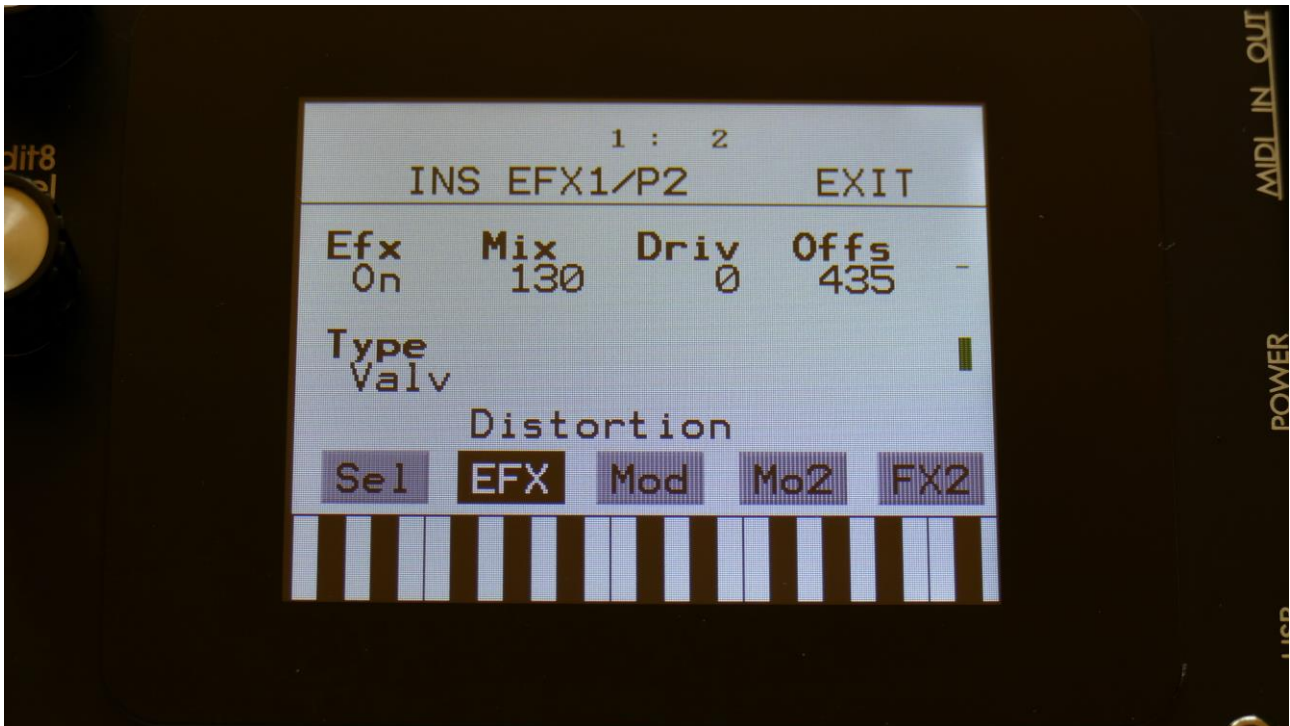
The parameters on this page:

Feed1 and Feed2: Modulates the chorus feedback.

Time: Modulates the time parameter.

Deep: Modulates the Deep parameter.

Distortion



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

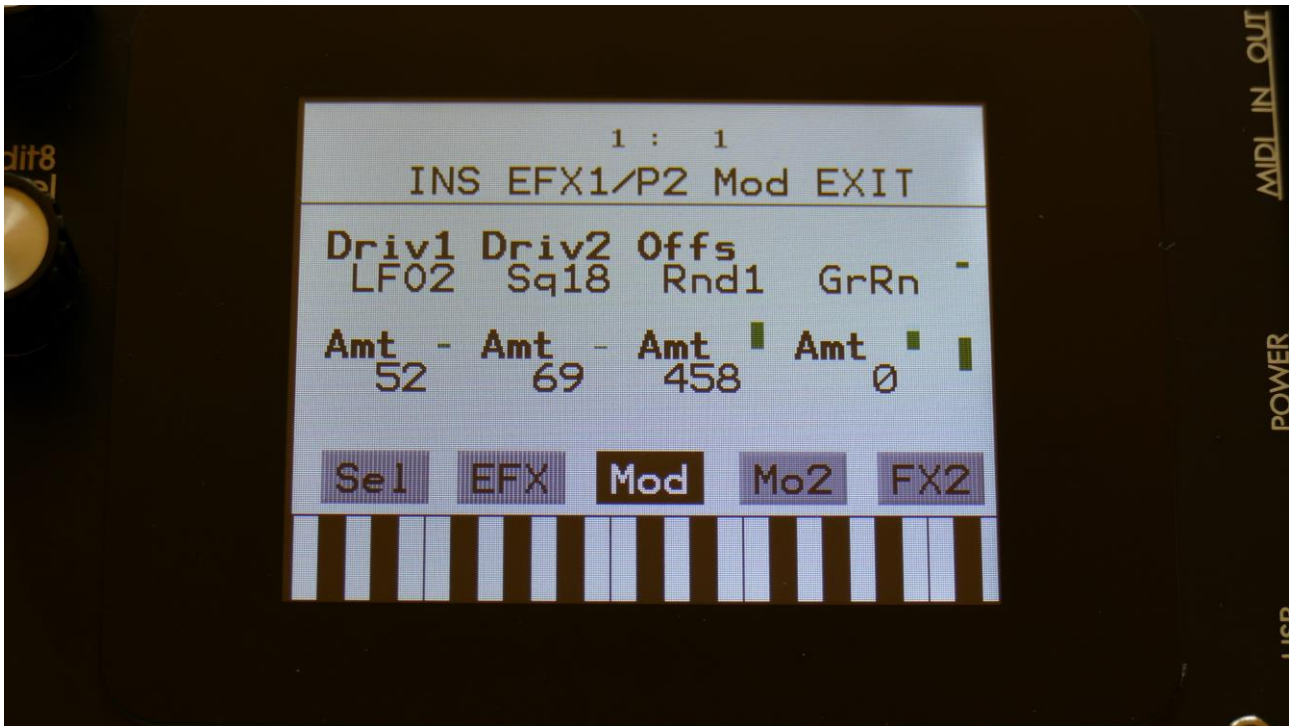
Drive: The higher the value, the more the sound will distort. If this is set to zero, no sound will pass through the distortion.

Offs: Distortion offset. The more this is turned up, the more asymmetric the distortion will get.

Type: Distortion type. Choices are:

- Valve:** A digital simulation of a classic valve distortion.
- Sine:** A noisy and warm sine shaping distortion.
- Fuzz:** Simulates a classic fuzz distortion.
- Xdis:** Complete destruction of the sound.

Distortion Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

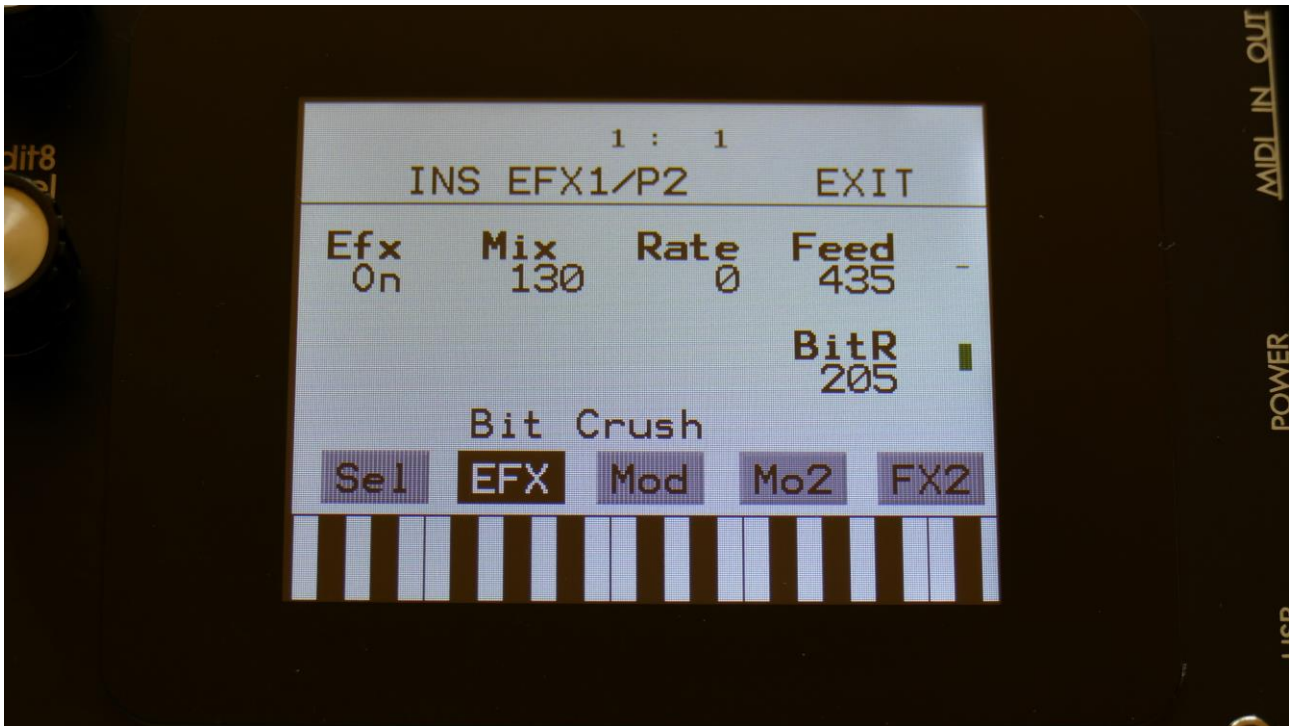
The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Driv1 and Driv2: Modulates the drive parameter.

Offs: Modulates the offset parameter.

Bit Crush



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

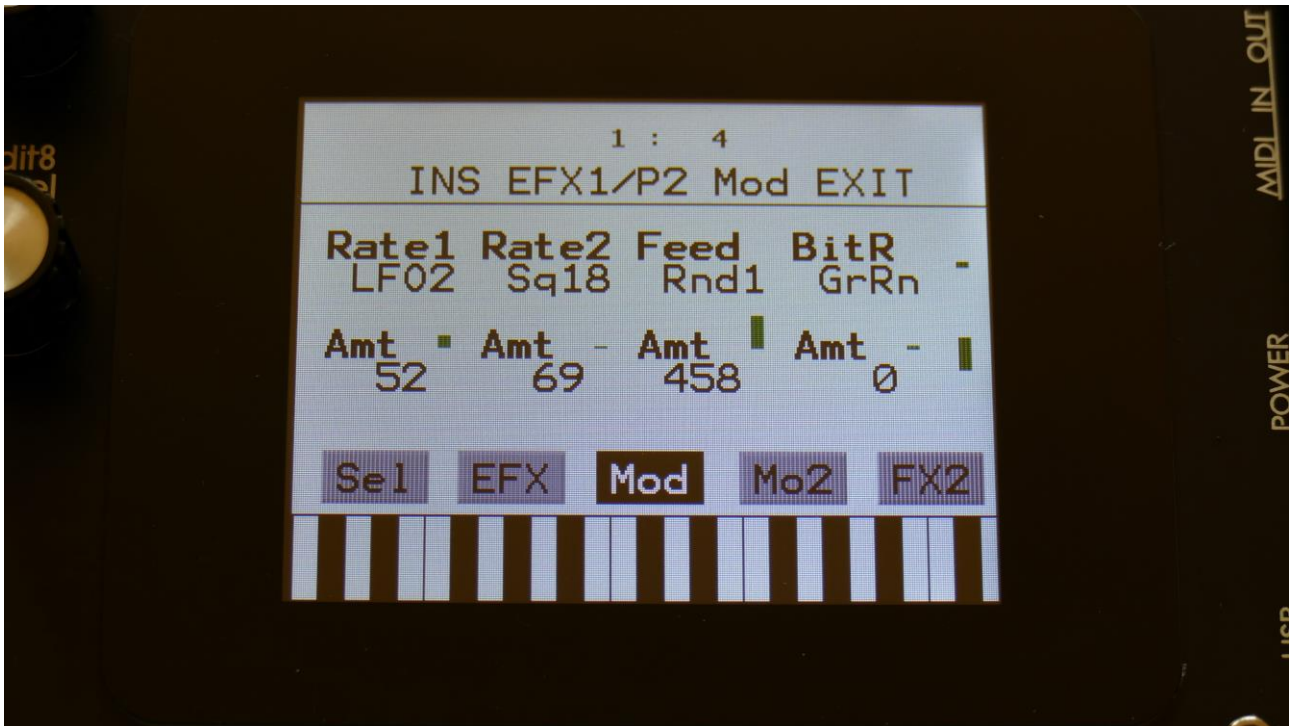
Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Rate: Sample Rate Reduction. The more this is turned up, the lower the sample rate will be. From 44.1 KHz to 1 KHz.

Feed: Feedback. Turning this up will slightly overdrive the sound.

BitR: Bit Reduction. The more this is turned up, the lower the bit resolution will get. When it is turned fully down, resolution is 16 bit, when turned fully up, it is 1 bit.

Bit Crush Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

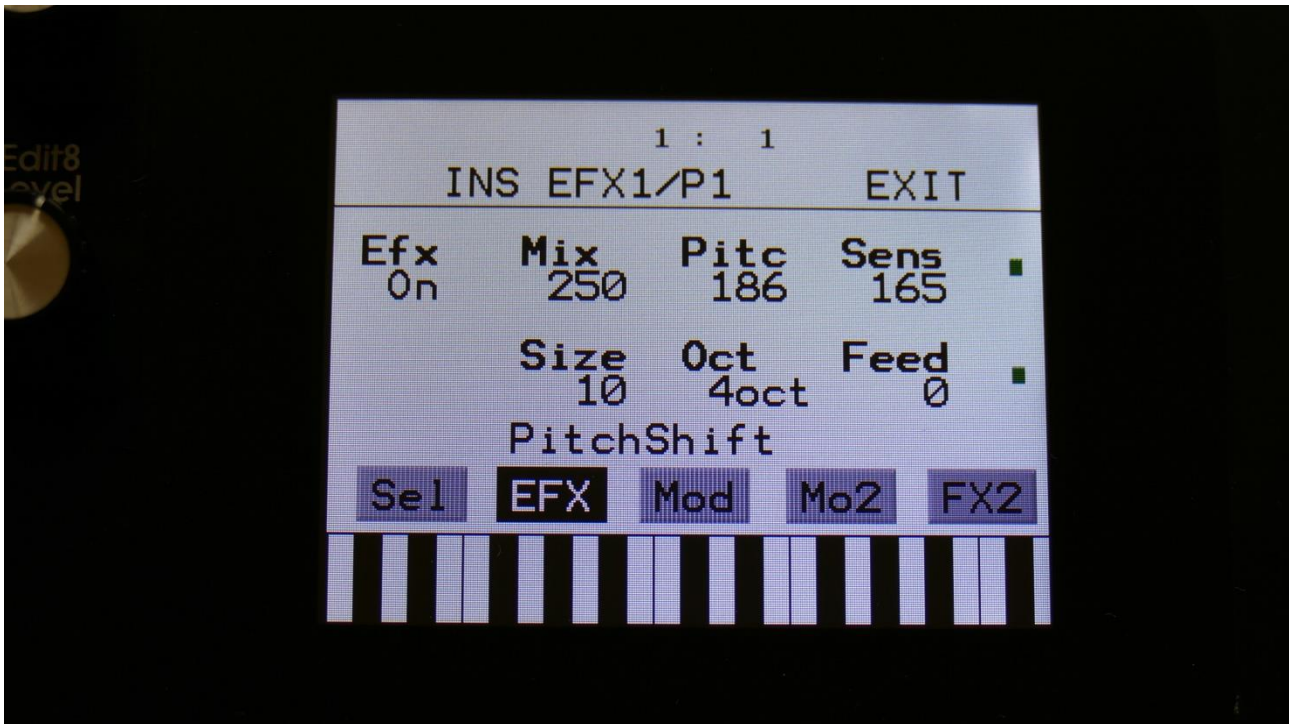
The parameters on this page:

Rate1 and Rate2: Modulates the sample rate parameter.

Feed: Modulates the feedback parameter.

BitR: Modulates the bit reduction parameter.

Pitch Shifter



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Pitc: Smoothly pitches the sound from up to 4 octaves below the original pitch, to 4 octaves above.

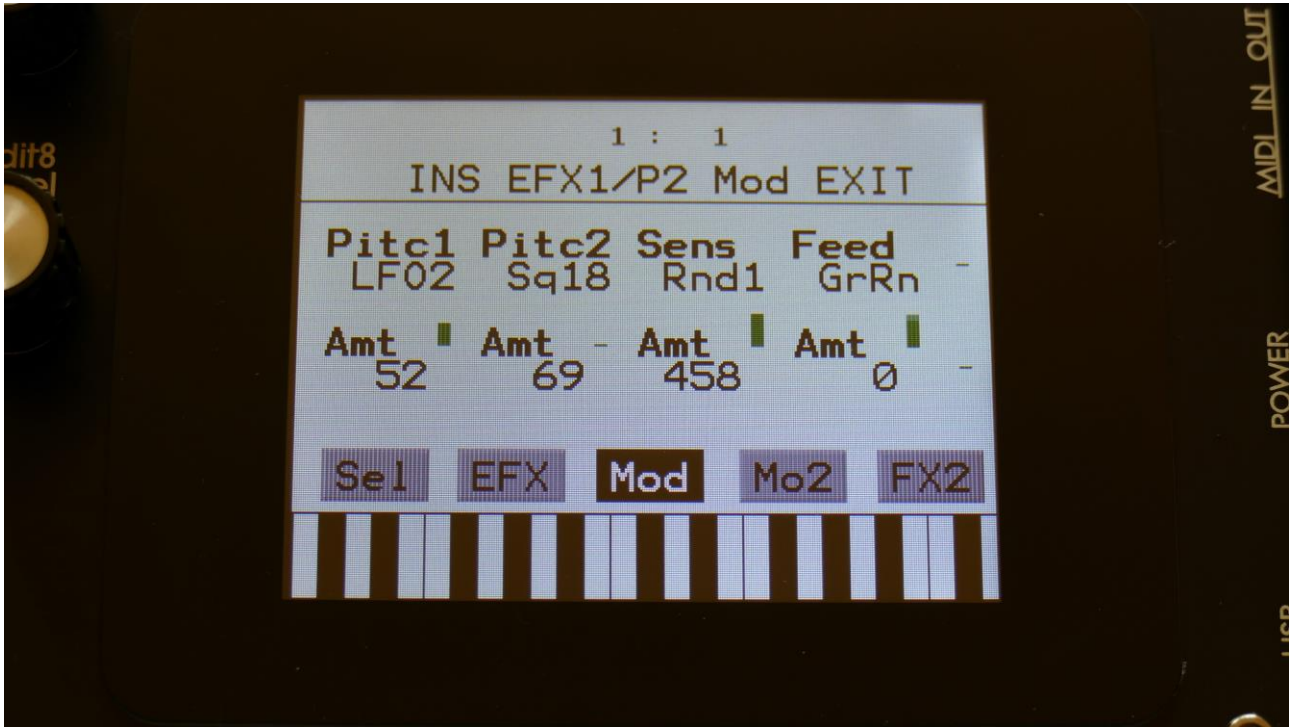
Sense: Pitch detection sense. On a pure waveform, turn this fully down to make sure, that it detects all the waves of it, and pitch shifts correctly. On more complex sounds, turn this up until the desired effect are obtained. At higher settings, only portions of the sound will be pitch shifted, and when it doesn't detect any pitch, it will repeat the portion it detected, making the sound "granulate".

Size: 1 to 64. This sets the size of the buffer, used for the pitch shifting effect. When pitch shifting more complex sounds, it might make the pitch shifting more precise, when turning this parameter up a bit.

Oct: The octave range of the pitch shifter. From +/- 1 to +/- 4 octaves.

Feed: Pitch shifter feedback. Adjusts the portion of the output signal, that is fed back to the input.

Pitch Shifter Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

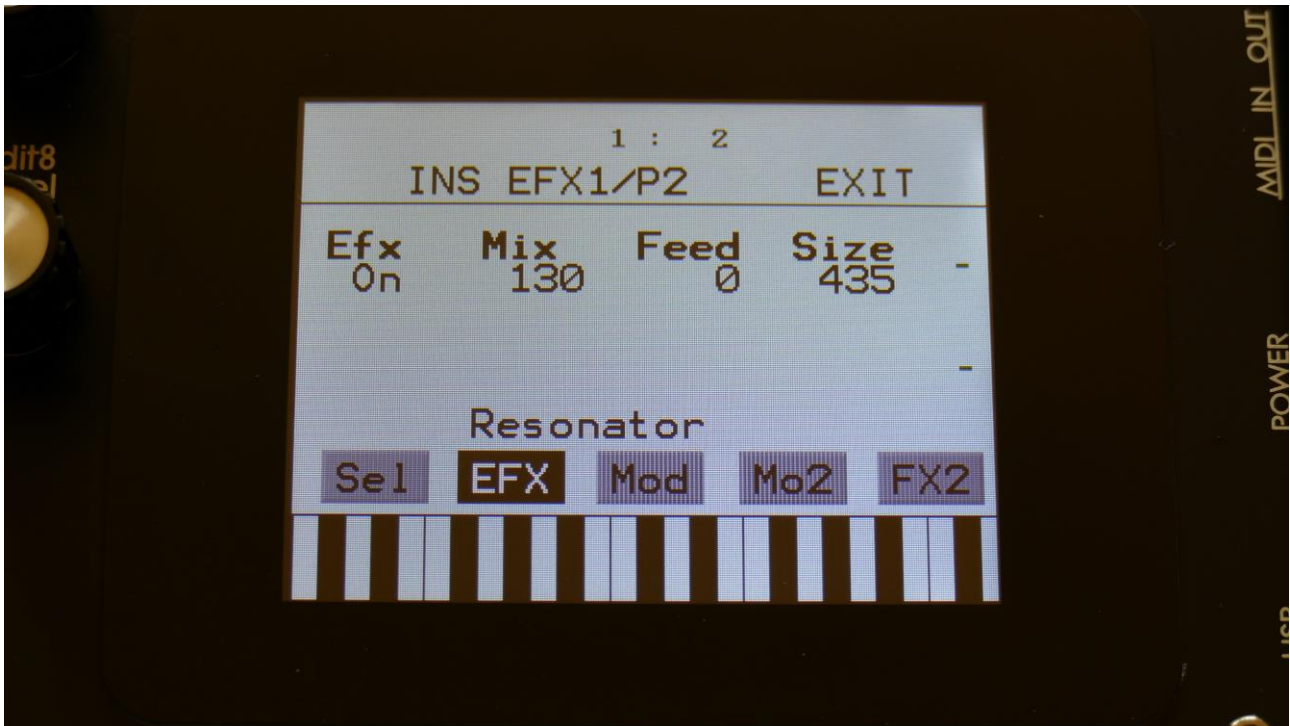
The parameters on this page:

Pitc1 and Pitc2: Modulates the pitch shift parameter.

Sens: Modulates the Sense parameter.

Feed: Modulates the Feed parameter.

Resonator



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

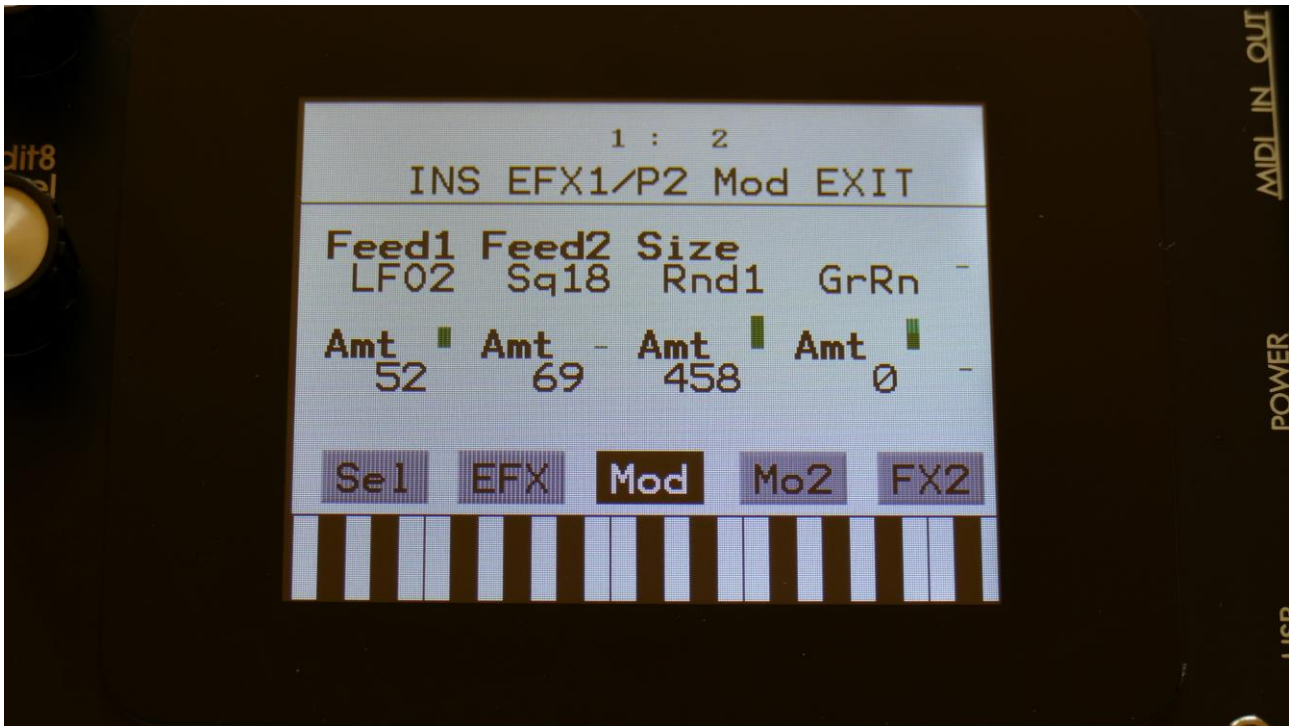
Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Feed: Resonator feedback. The more this is turned up, the more it will resonate.

Size: The size of the resonator box. Different sizes will give different resonance frequencies.

Resonator Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

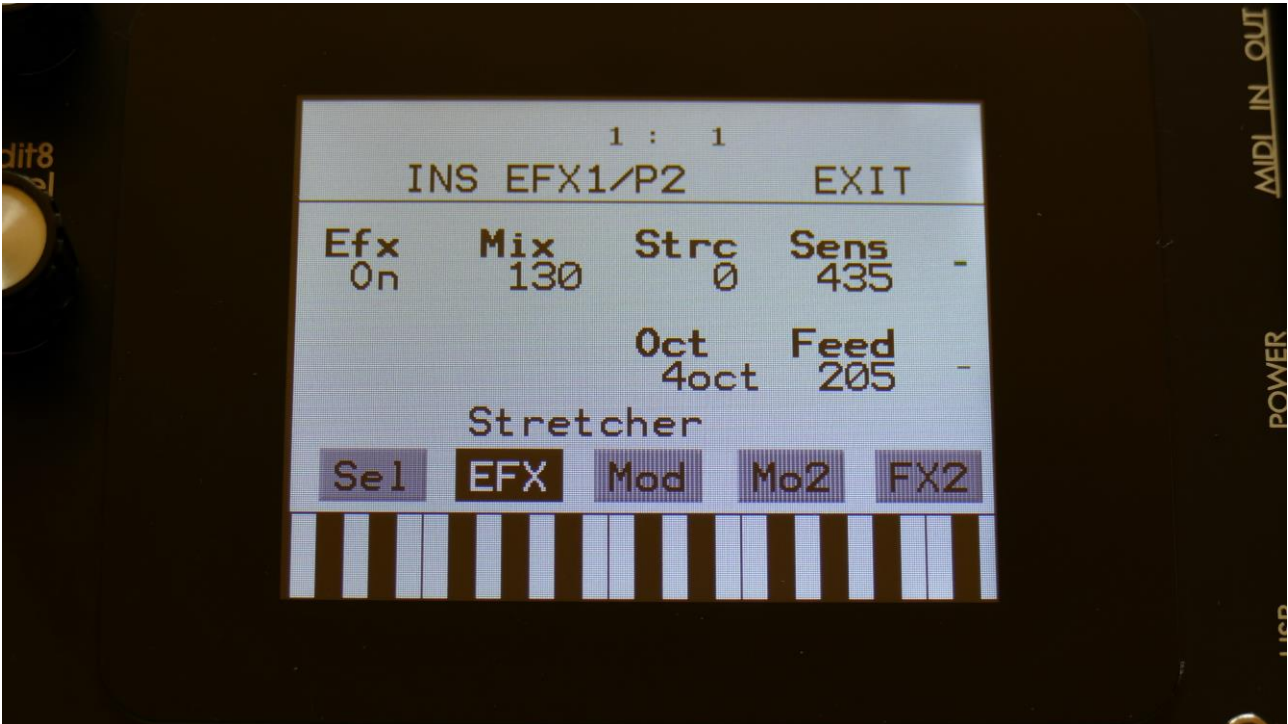
The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Feed1 and Feed2: Modulates the feed parameter.

Size: Modulates the Size parameter.

Stretcher



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

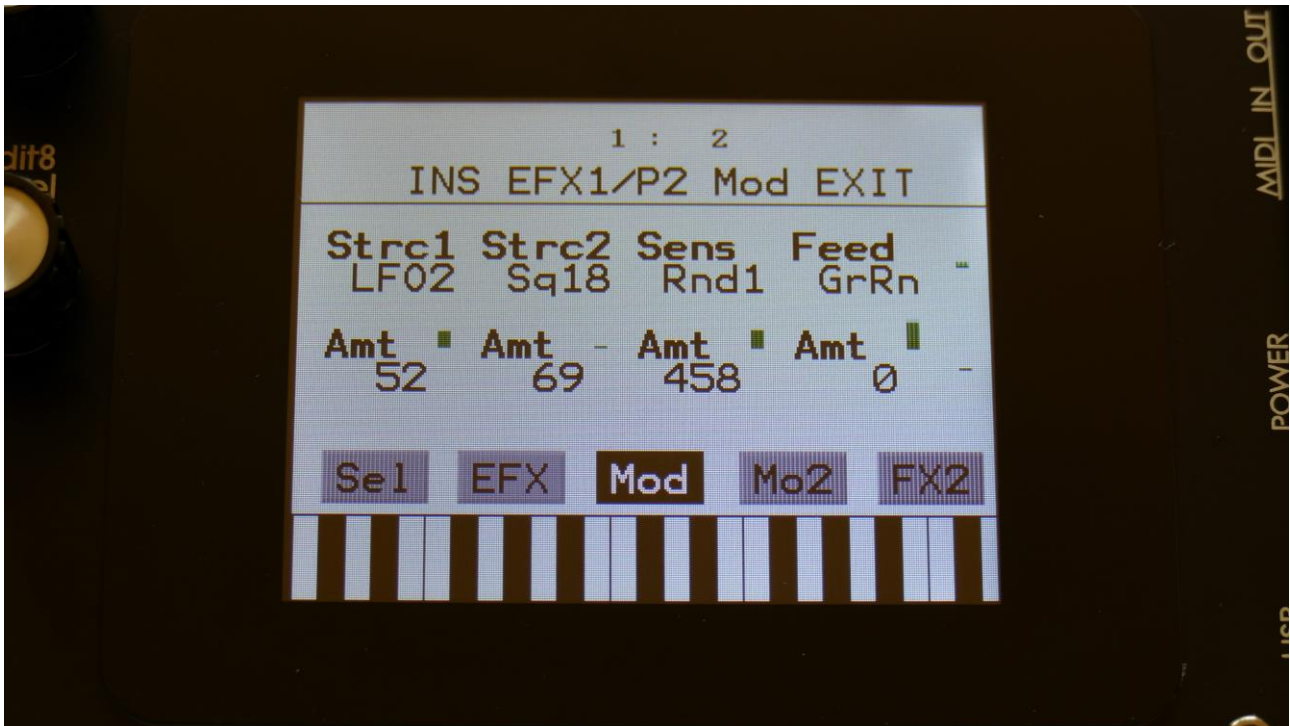
Strc: The degree of time stretch.

Sens: Stretch detection sense. At lower settings the sound will “wobble”, at higher settings it will “granulate”. Adjust this to obtain different effects.

Oct: The octave range of the stretch effect. From +/- 1 to +/- 4 octaves.

Feed: Stretcher feedback. Adjusts the portion of the output signal, that is fed back to the input.

Stretcher Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

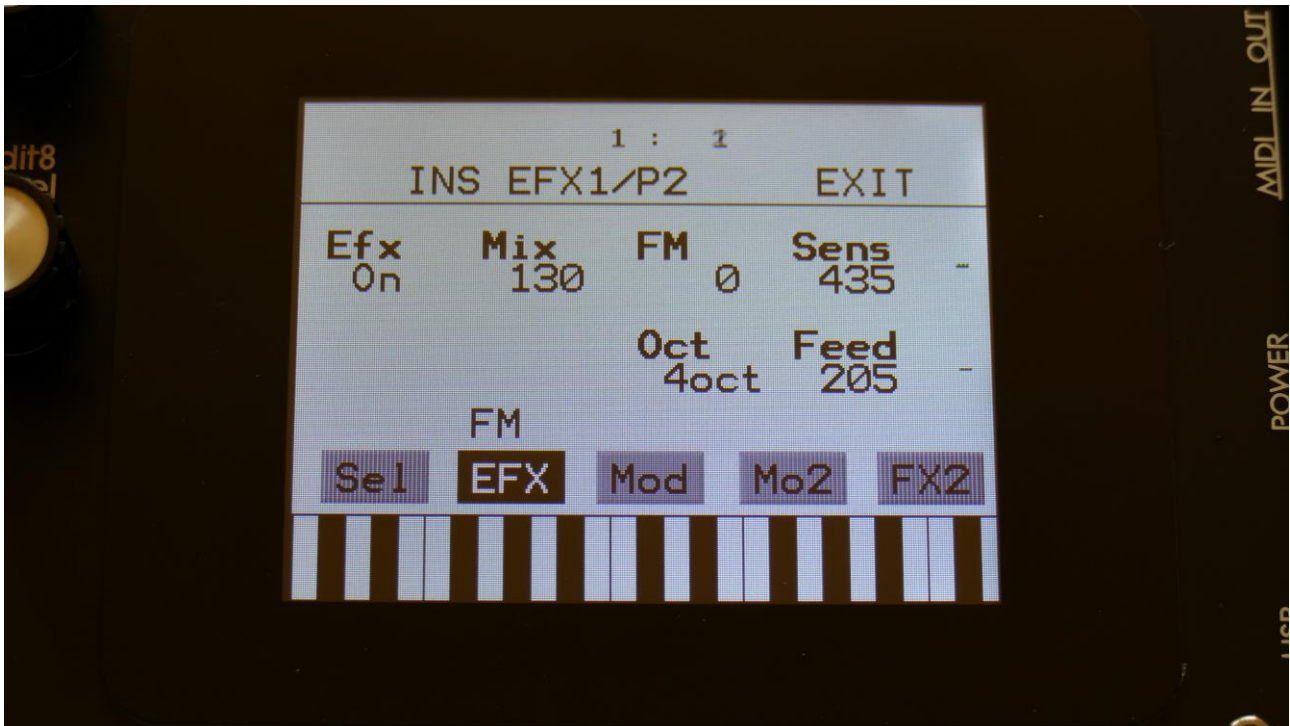
The parameters on this page:

Strc1 and Strc2: Modulates the stretch parameter.

Sens: Modulates the sense parameter.

Feed: Modulates the feed parameter.

FM



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

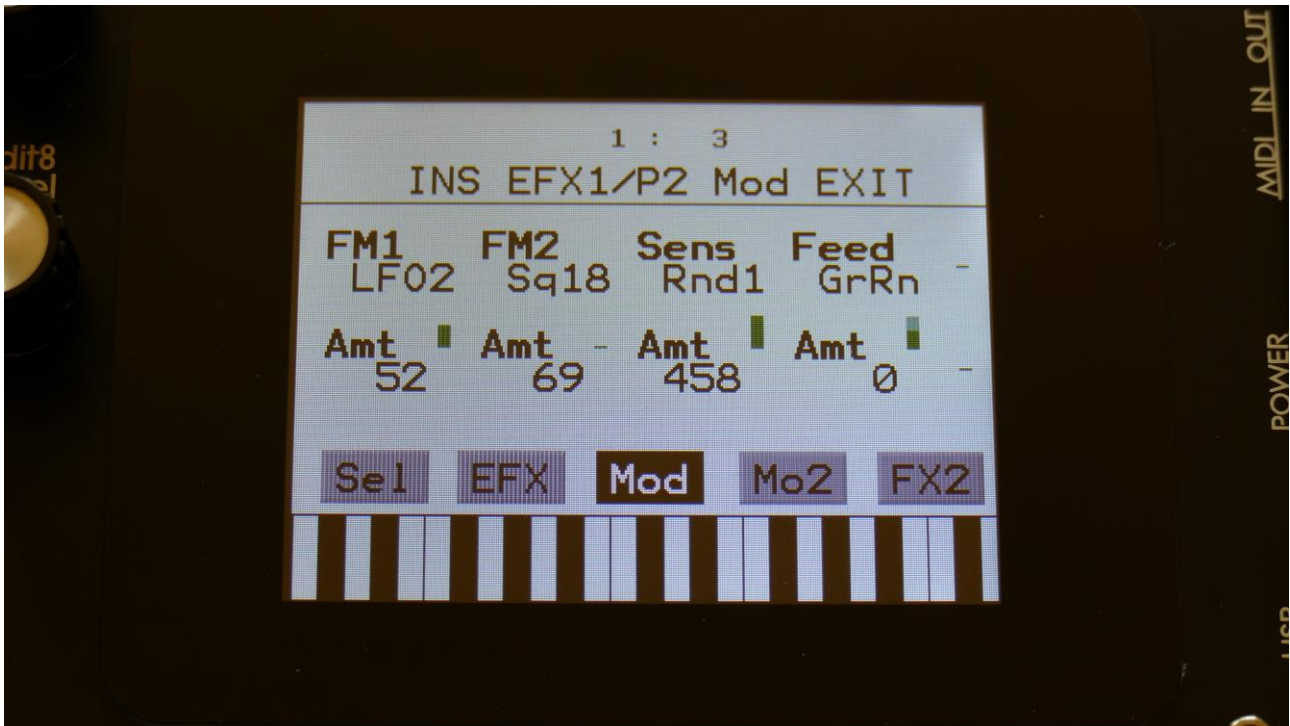
FM: The amount of self-FM from 0 to up to +/- 4 octaves.

Sense: FM pitch detection sense. On a pure waveform, turn this fully down to make sure, that it detects all the waves of it, and pitch shifts correctly. On more complex sounds, turn this up until the desired effect are obtained. At higher settings, only portions of the sound will be pitch shifted, and when it doesn't detect any pitch, it will repeat the portion it detected, making the sound "granulate".

Oct: The octave range of the FM effect. From +/- 1 to +/- 4 octaves.

Feed: FM feedback. Adjusts the portion of the output signal, that is fed back to the input.

FM Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

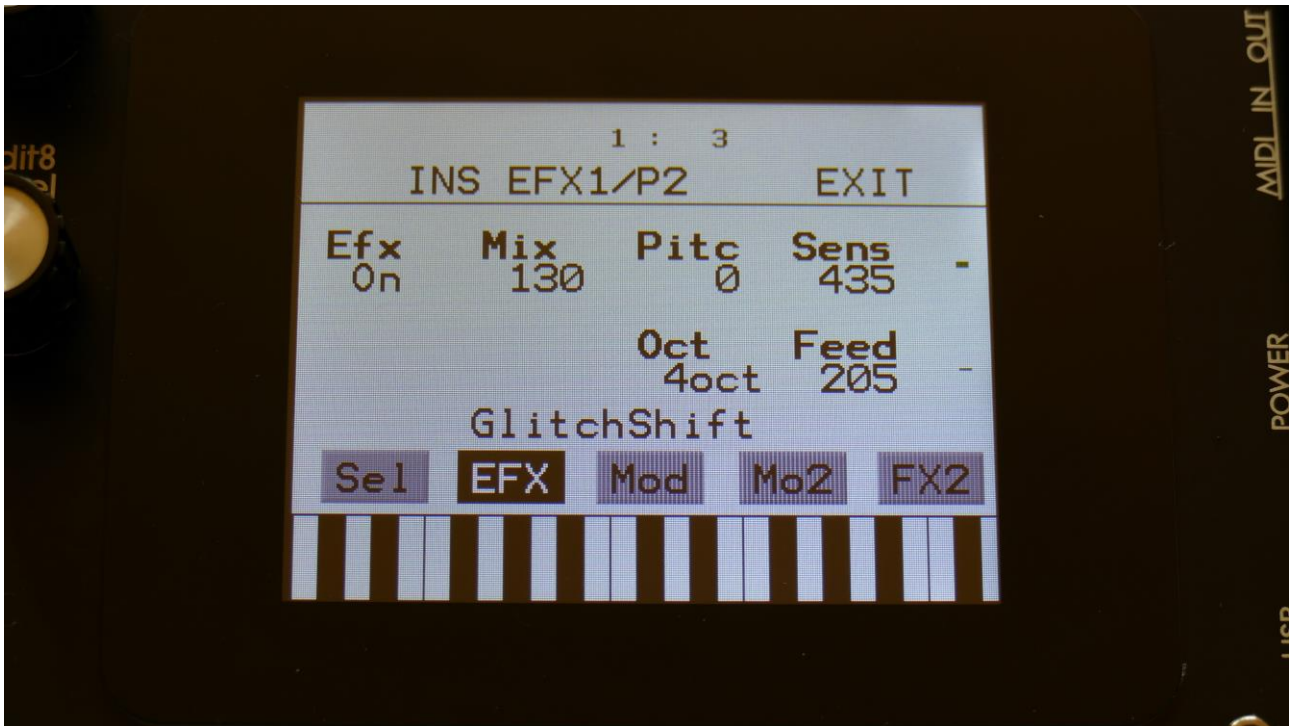
The parameters on this page:

FM1 and FM2: Modulates the FM parameter.

Sens: Modulates the sense parameter.

Feed: Modulates the feed parameter.

Glitch Shifter, Glitch Shifter 2, Glitch Shifter 3



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

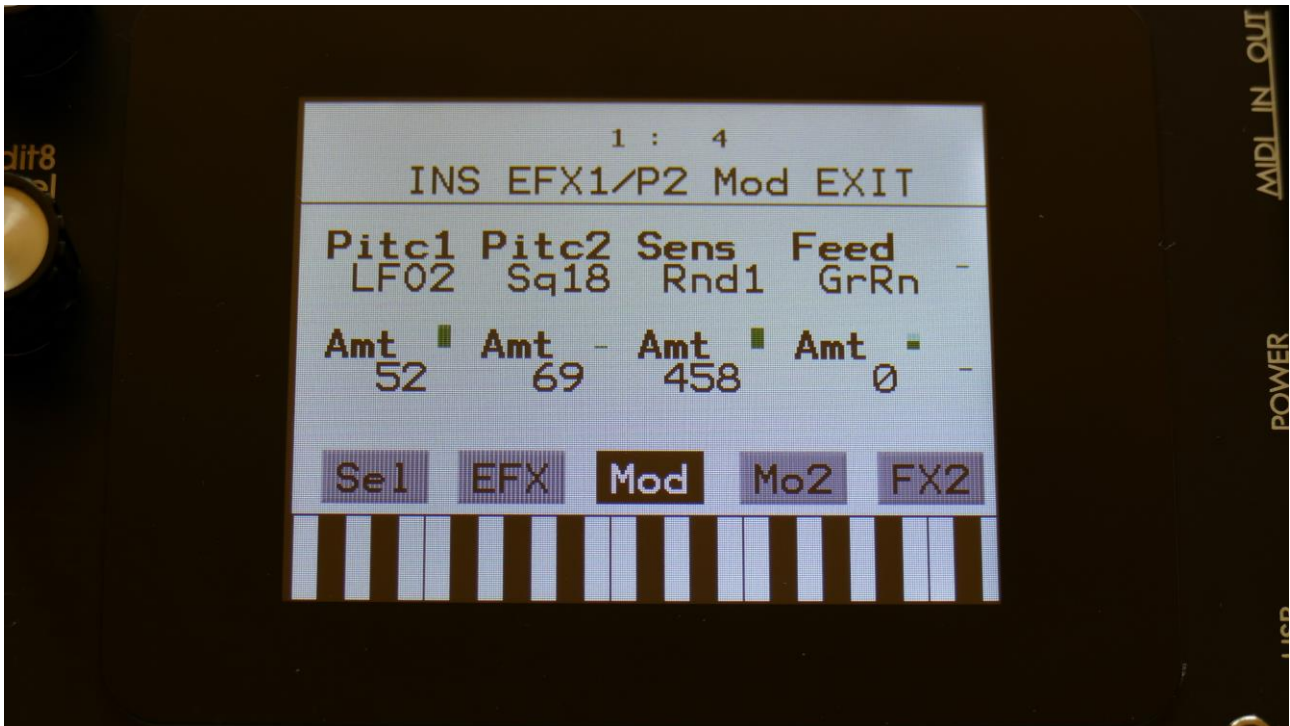
Pitc: Glitchy pitches the sound from up to 4 octaves below the original pitch, to 4 octaves above.

Sense: Pitch detection sense. On a pure waveform, turn this fully down to make sure, that it detects all the waves of it, and pitch shifts correctly. On more complex sounds, turn this up until the desired effect are obtained. At higher settings, only portions of the sound will be pitch shifted, and when it doesn't detect any pitch, it will repeat the portion it detected, making the sound "granulate".

Oct: The octave range of the glitch shifter. From +/- 1 to +/- 4 octaves.

Feed: Glitch shifter feedback. Adjusts the portion of the output signal, that is fed back to the input.

Glitch Shifter, Glitch Shifter 2, Glitch Shifter 3 Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

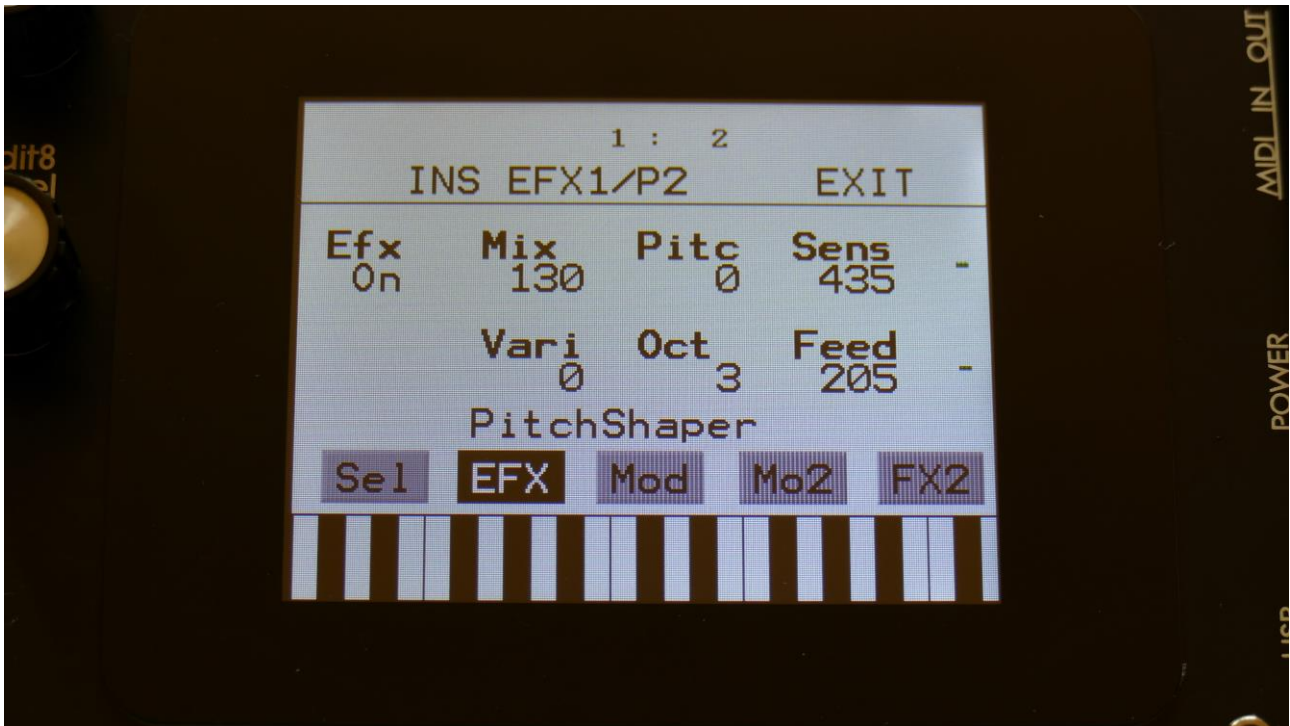
The parameters on this page:

Pitc1 and Pitc2: Modulates the glitch shift parameter.

Sens: Modulates the Sense parameter.

Feed: Modulates the Feed parameter.

Pitch Shaper



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Pitc: Sets the frequency, that the input signal should be re-pitched to.

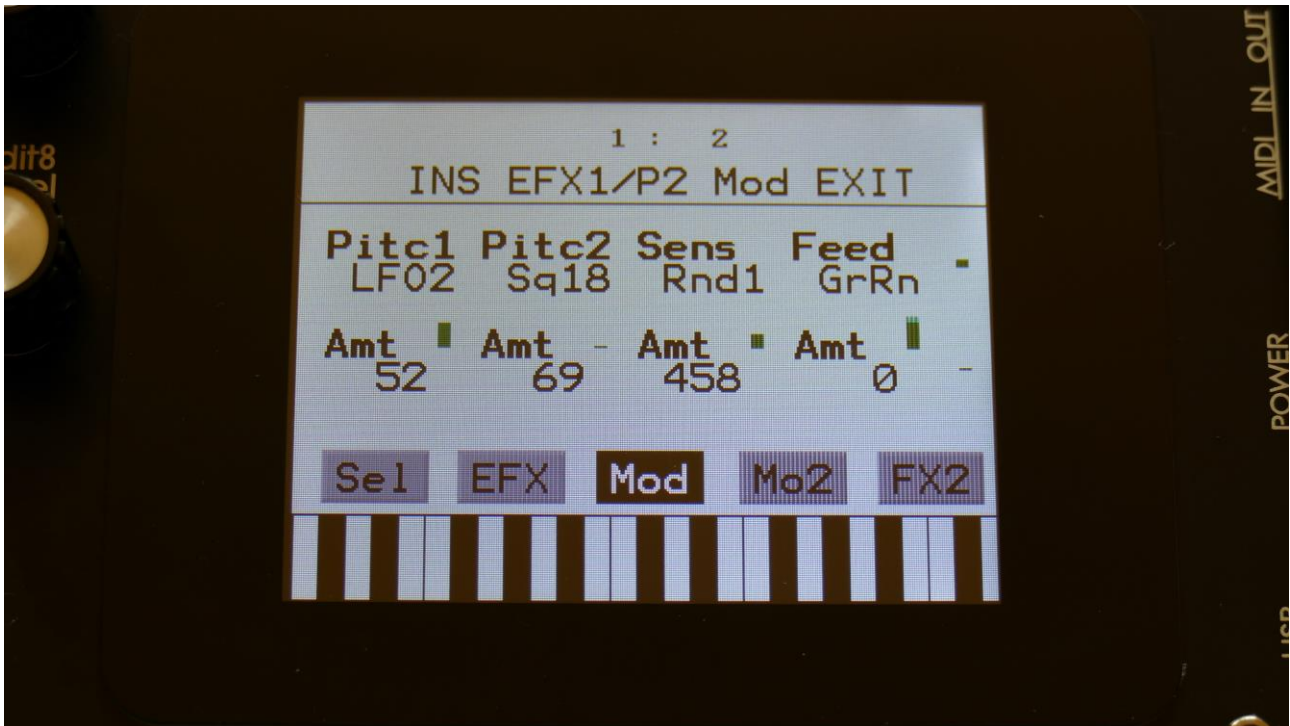
Sense: Pitch detection sense. On a pure waveform, turn this fully down to make sure, that it detects all the waves of it, and pitch shifts correctly. On more complex sounds, turn this up until the desired effect are obtained. At higher settings, only portions of the sound will be pitch shifted, and when it doesn't detect any pitch, it will repeat the portion it detected, making the sound "granulate".

Vari: Pitch variation. The more this is turned up, the more the pitch variations on the input signal affects the pitch shaper frequency.

Oct: Octave transpose. From 0 to +3 octaves.

Feed: Pitch shaper feedback. Adjusts the portion of the output signal, that is fed back to the input.

Pitch Shaper Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

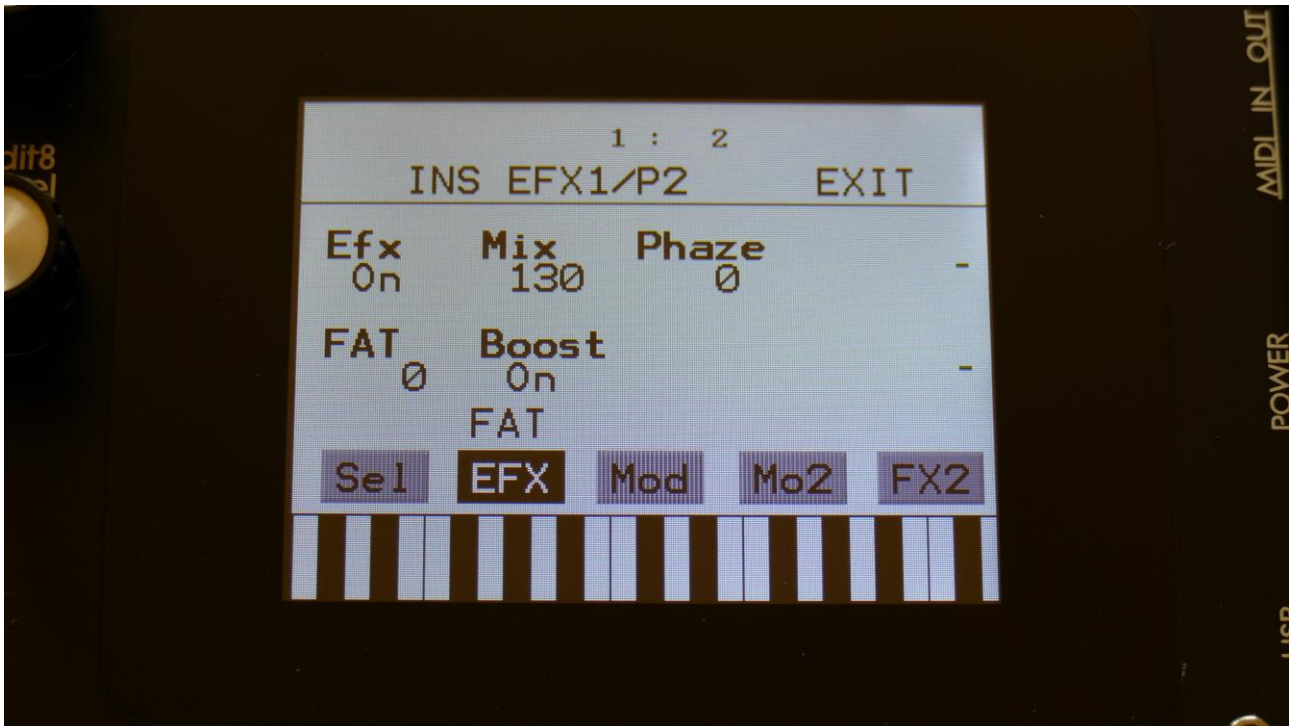
The parameters on this page:

Pitc1 and Pitc2: Modulates the pitc parameter.

Sens: Modulates the Sense parameter.

Feed: Modulates the Feed parameter.

FAT



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

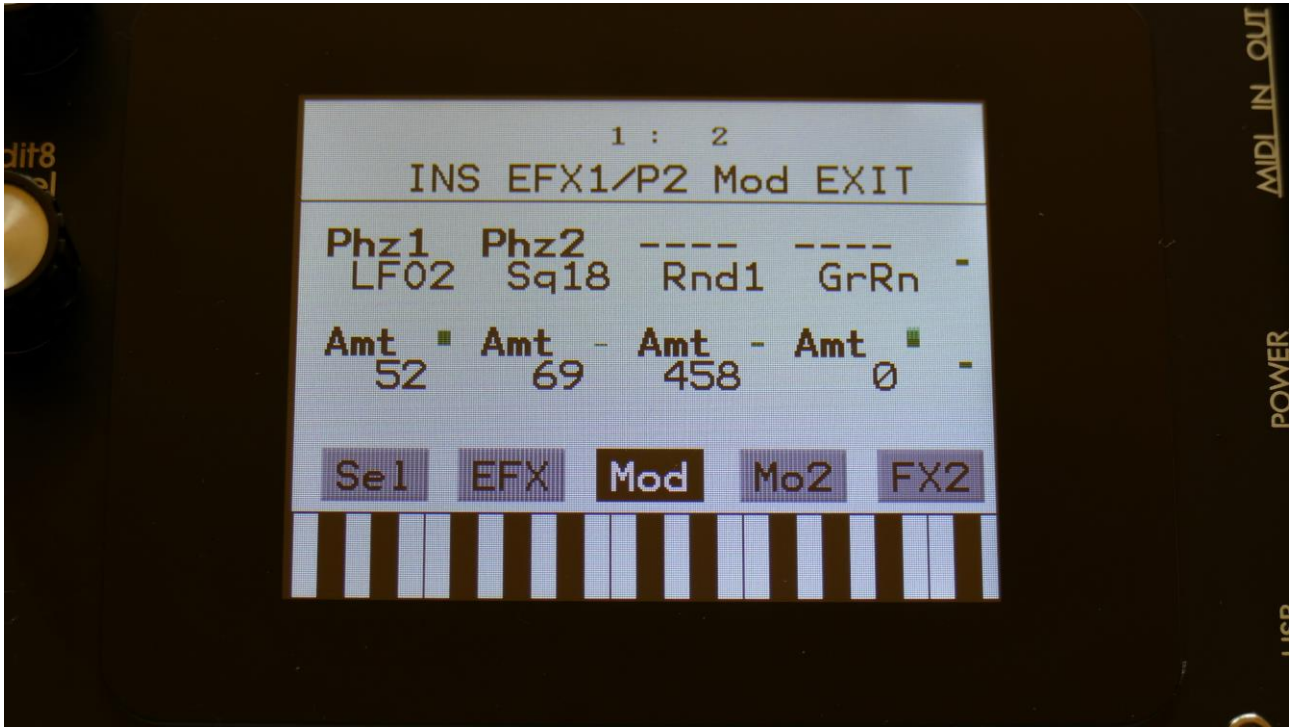
Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Phase: Adjusts the phase difference of the FAT layers. When turned fully down, all layers are in phase. The more it is turned up, the more the layers goes out of phase to each other. Useful for creating comb filter type sounds.

FAT: Adds 0 to 3 layers of the input sound to itself.

Boost: When this is on, the FAT layers are sonically added to each other, which will make the sound louder, and might cause it to saturate. If it is off, the sound stays at its original sonic level, even when layers are added.

FAT Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

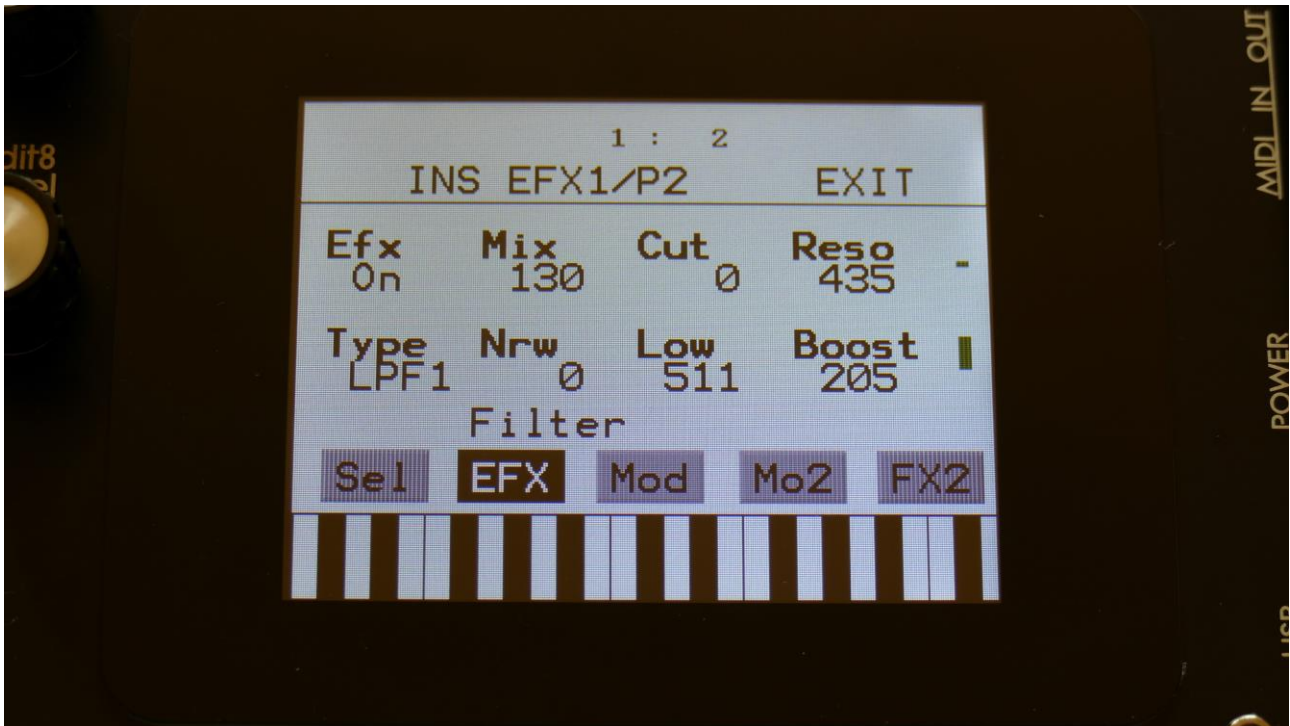
For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Phz1 and Phz2: Modulates the Phaze parameter.

Filter



The filter types:

LPF1: Lowpass filter with a rather weak character. Resonance does not self-oscillate.

LPF2: Lowpass filter that are a bit sharper than LPF1. Resonance does not self-oscillate.

LPF3: Sharp Lowpass filter with self-oscillating resonance.

LPF4: Very sharp Lowpass filter with self-oscillating resonance.

BPF1: Bandpass filter with a rather weak character. Resonance does not self-oscillate.

BPF2: Bandpass filter with focus on the bass area. Distorts at higher input levels. Resonance does not self-oscillate.

BPF3: Sharp Bandpass filter with self-oscillating resonance.

HPF1: Sharp Highpass filter with self-oscillating resonance.

HPF2: High gained Highpass filter with self-oscillating resonance. Distorts at higher input levels.

HPF3: Sharp Highpass filter with self-oscillating resonance. A bit weaker than HPF1.

Dstr: Destruktion. A rather defective filter. Self-oscillates at some points, distorts at others.

FAT1: Slightly distorting lowpass filter, with a sharp response and self-oscillating resonance.

FAT2: A slightly weaker version of FAT1.

LoFi: A very distorting and unpredictable lowpass filter.

LPF5: A lowpass filter with a very soft character. Resonance does not self-oscillate.

BPF4: A bandpass filter with a very soft character and self-oscillating resonance.

Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Cut: 0 to 511. Sets the cutoff frequency of the filter.

Reso: 0 to 511. Sets the amount of resonance applied to the filter.

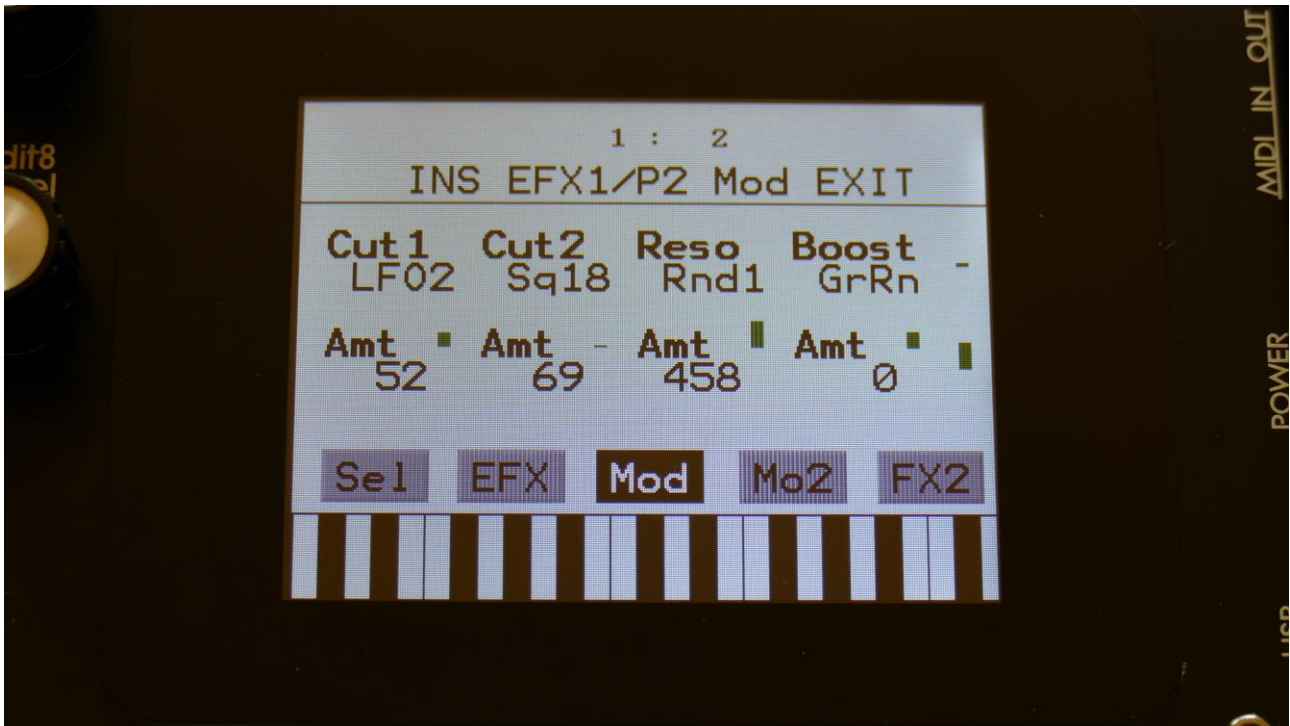
Type: Sets the filter type. See the list of filter types.

Nrw: 0 to 511. Turning this up, will make the frequency response of the filter more and more narrow.

Low: 0 to 511. Sets the lowest frequency offset point of the filter. Turning this up, will in many filter types, make the bass bottom more present or distorted.

Boost: 0 to 511. Gains the filter output level.

Filter Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

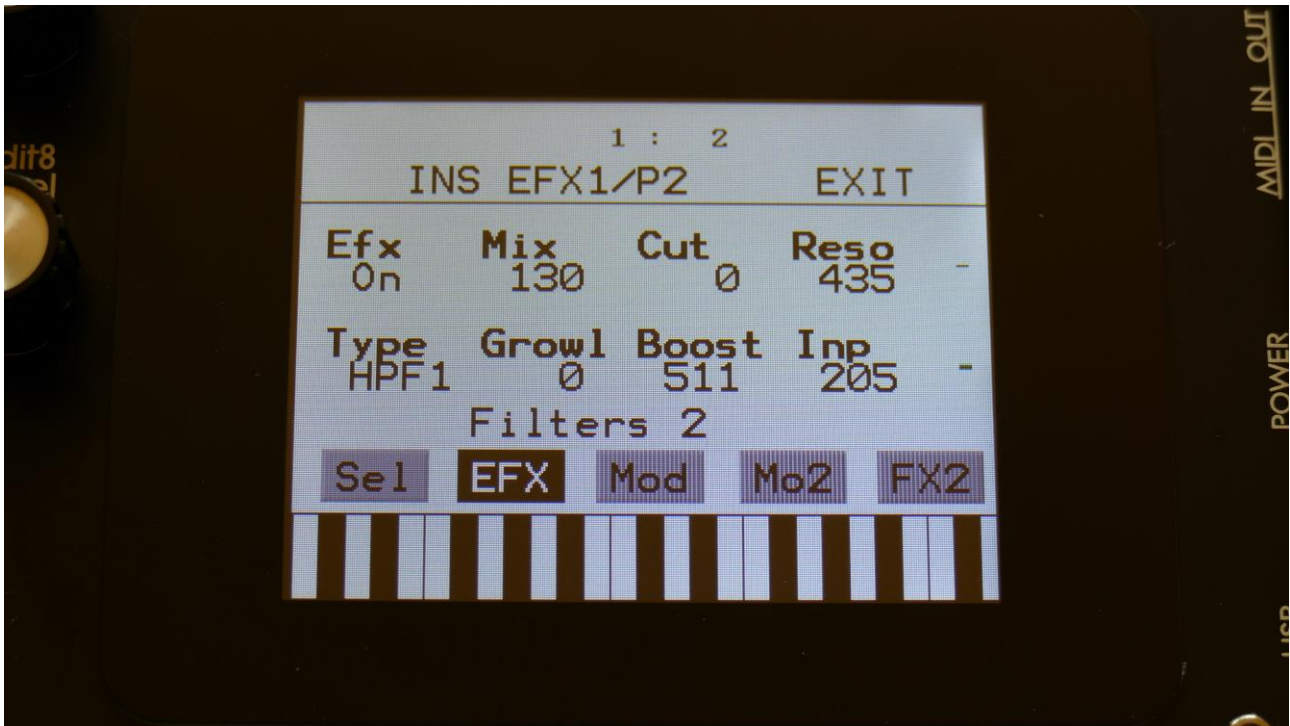
The parameters on this page:

Cut1, Cut2: Will modulate the filter cutoff frequency.

Reso: Will modulate the amount of resonance applied to the filter.

Boost: Will modulate the filter output gaining.

Filters 2



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Cut: Filter cutoff frequency.

Reso: Filter resonance setting.

Type: Filter type. Possibilities are:

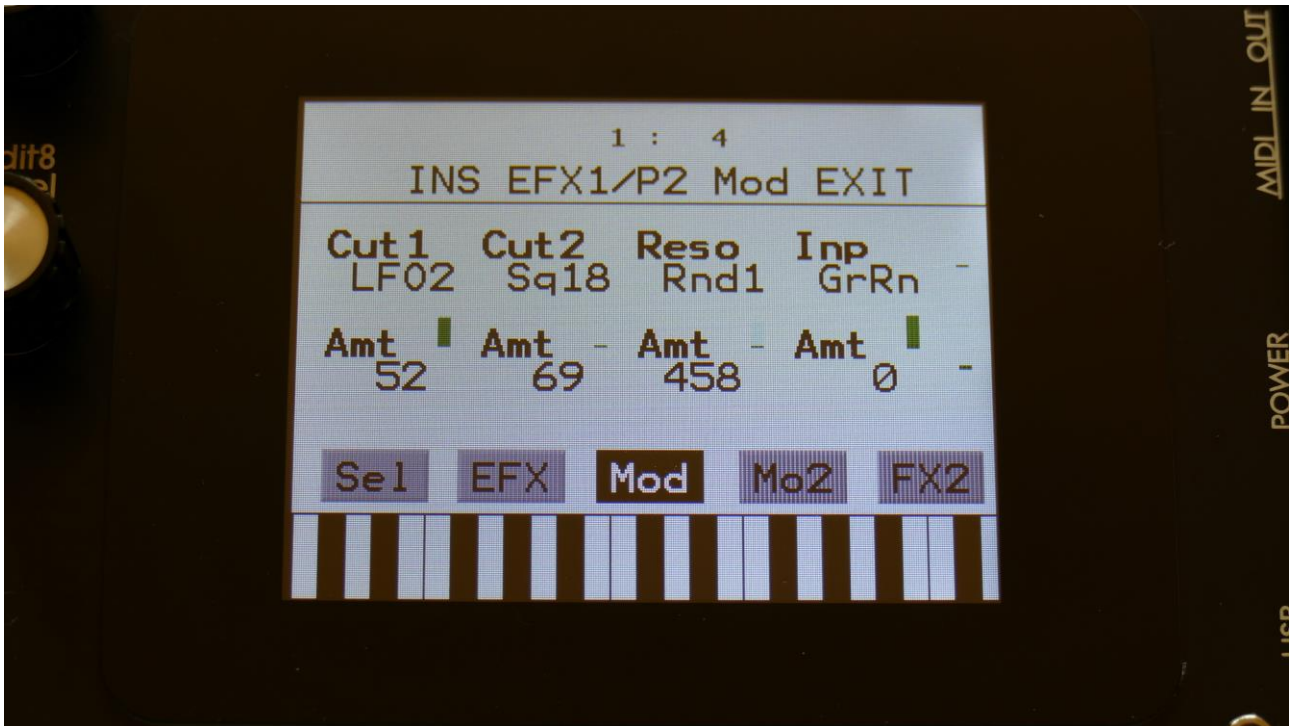
- HPF1:** High pass filter with extra bass bottom and uncontrollable resonance.
- LPF1:** Low pass filter with extra bass bottom and uncontrollable resonance.
- BPF1:** Band pass filter with extra bass bottom and uncontrollable resonance.
- HPF2:** High pass filter with a thinner and more resonant sound.
- LPF2:** Low pass filter with a thinner and more resonant sound.
- BPF2:** Band pass filter with a thinner and more resonant sound.

Growl: Adds unsymmetric distortion to the filter.

Boost: Adds extra boost to the filter output signal, if desired.

Inp: The input level of the filter.

Filters 2 Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

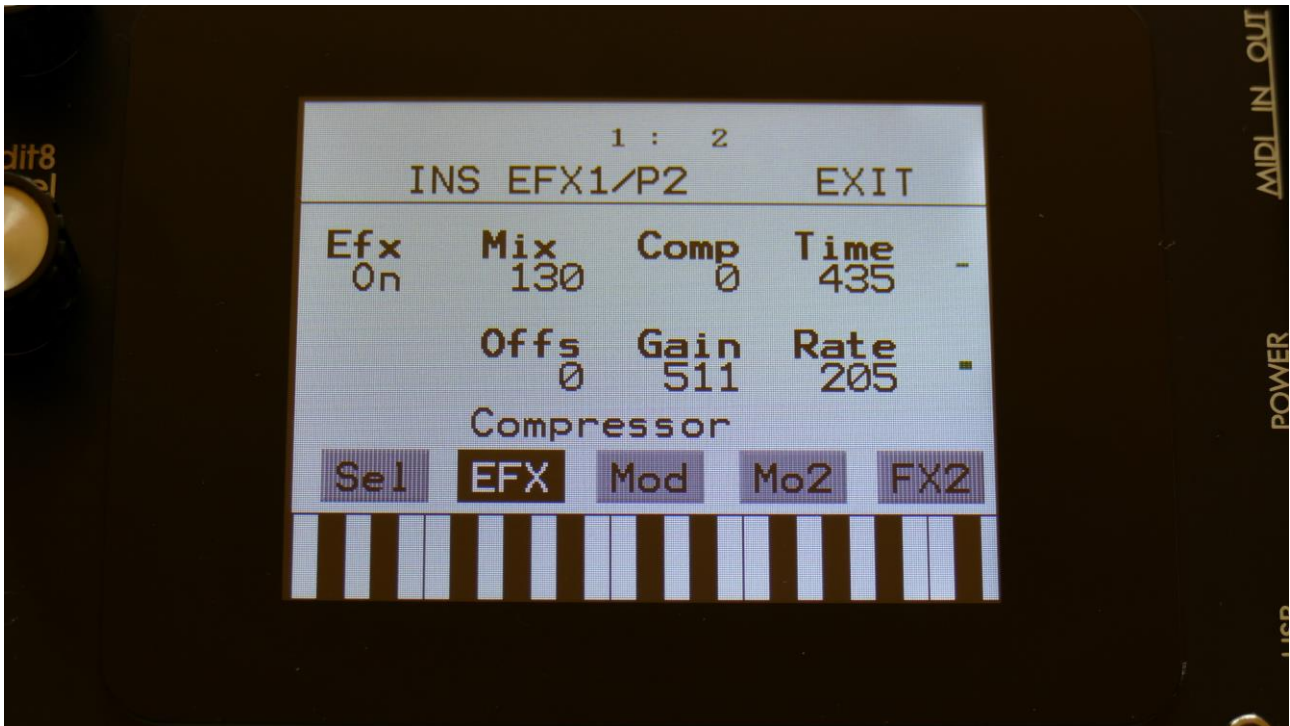
The parameters on this page:

Cut1, Cut2: Modulates the filter cutoff frequency.

Reso: Modulates the filter resonance.

Inp: Modulates the filter input level parameter.

Compressor



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Comp: 0 to 511. Compression amount. The more this is turned up, the more the portion of the signal, that are higher than the offset value, will be damped.

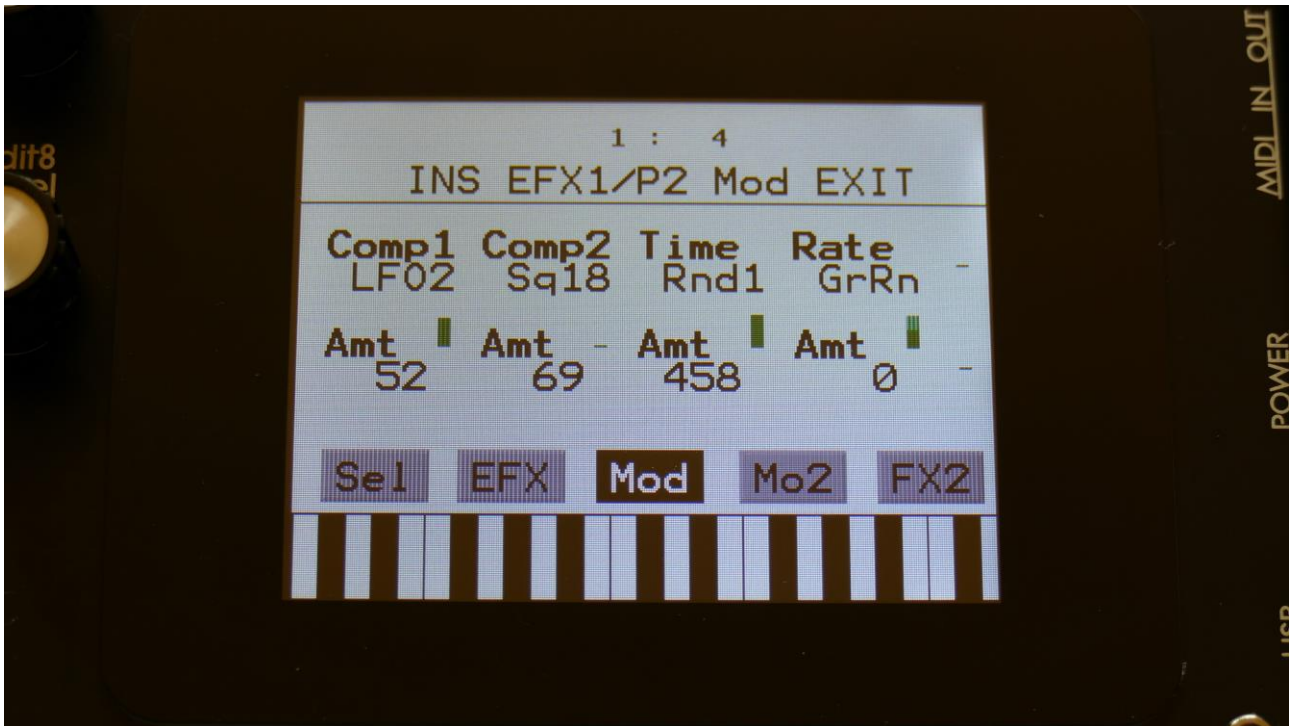
Offs: 0 to 511. Sets the level point where the compression will kick in. The more this is turned up, the higher the input signal will need to be, before it starts to compress.

Gain: 0 to 511. Sets the output gain of the compressor.

Time: 0 to 511. The more this is turned up, the longer time it will take, for the compression to kick in.

Rate: 0 to 511. This sets how fast the compressor will react to level changes of the input signal. The more it is turned up, the slower it will react.

Compressor Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

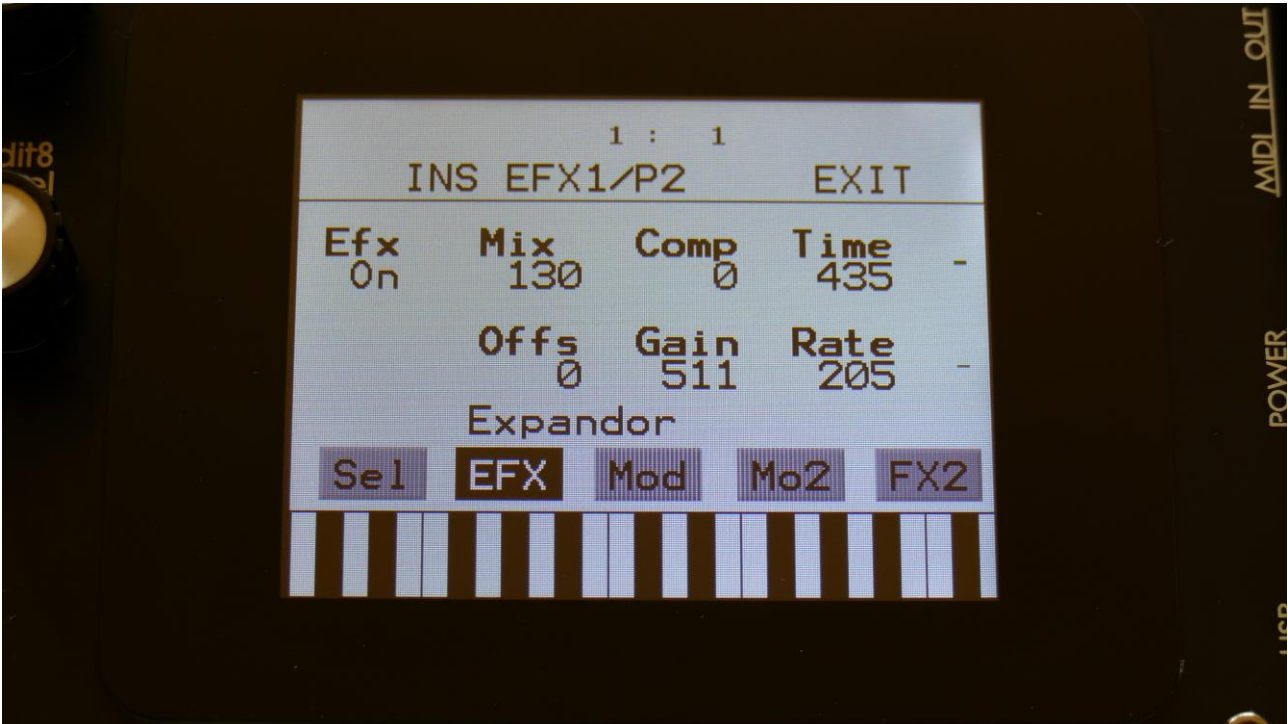
The parameters on this page:

Comp1, Comp2: Will modulate the compression amount.

Time: Will modulate the compression “kick in” time.

Rate: Will modulate the time it takes for the compressor to react to changes on the input signal.

Expander



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Comp: 0 to 511. Compression amount. The more this is turned up, the more the portion of the signal, that are higher than the offset value, will be gained.

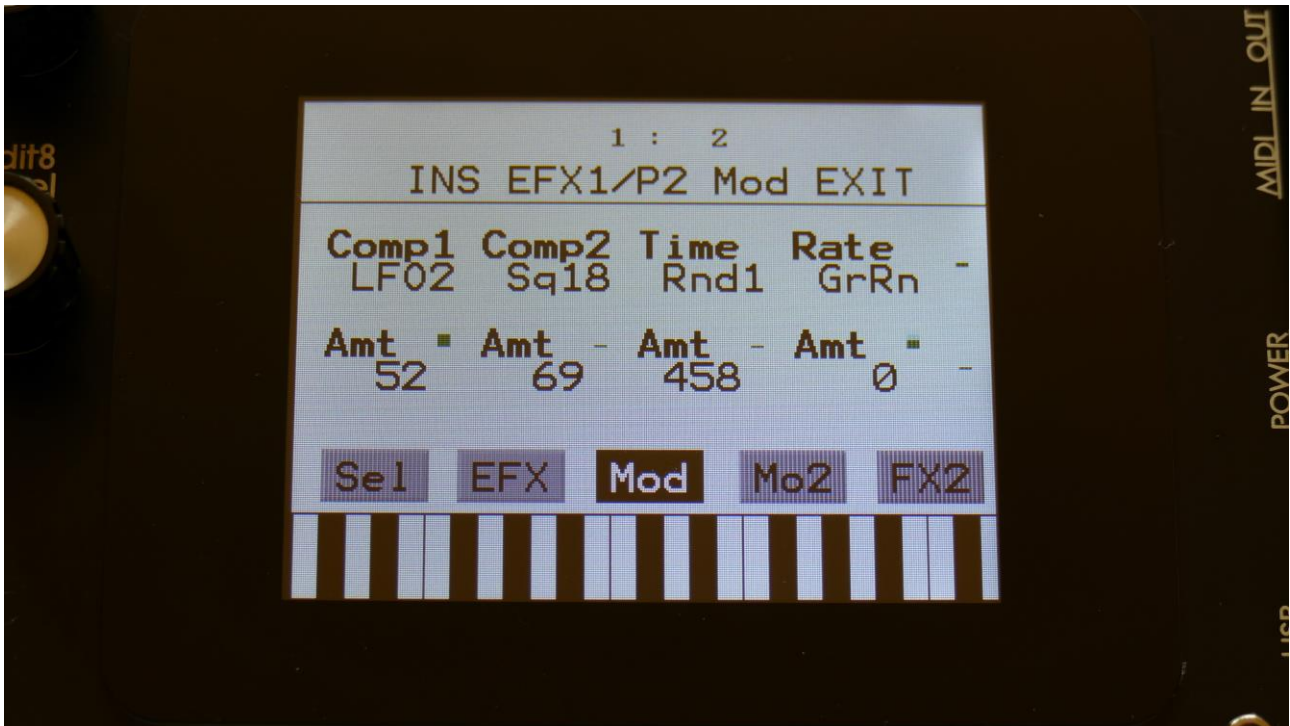
Offs: 0 to 511. Sets the level point where the expansion will kick in. The more this is turned up, the higher the input signal will need to be, before it starts to gain.

Gain: 0 to 511. Sets the output gain of the expander.

Time: 0 to 511. The more this is turned up, the longer time it will take, for the expansion to kick in.

Rate: 0 to 511. This sets how fast the expander will react to level changes of the input signal. The more it is turned up, the slower it will react.

Expander Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

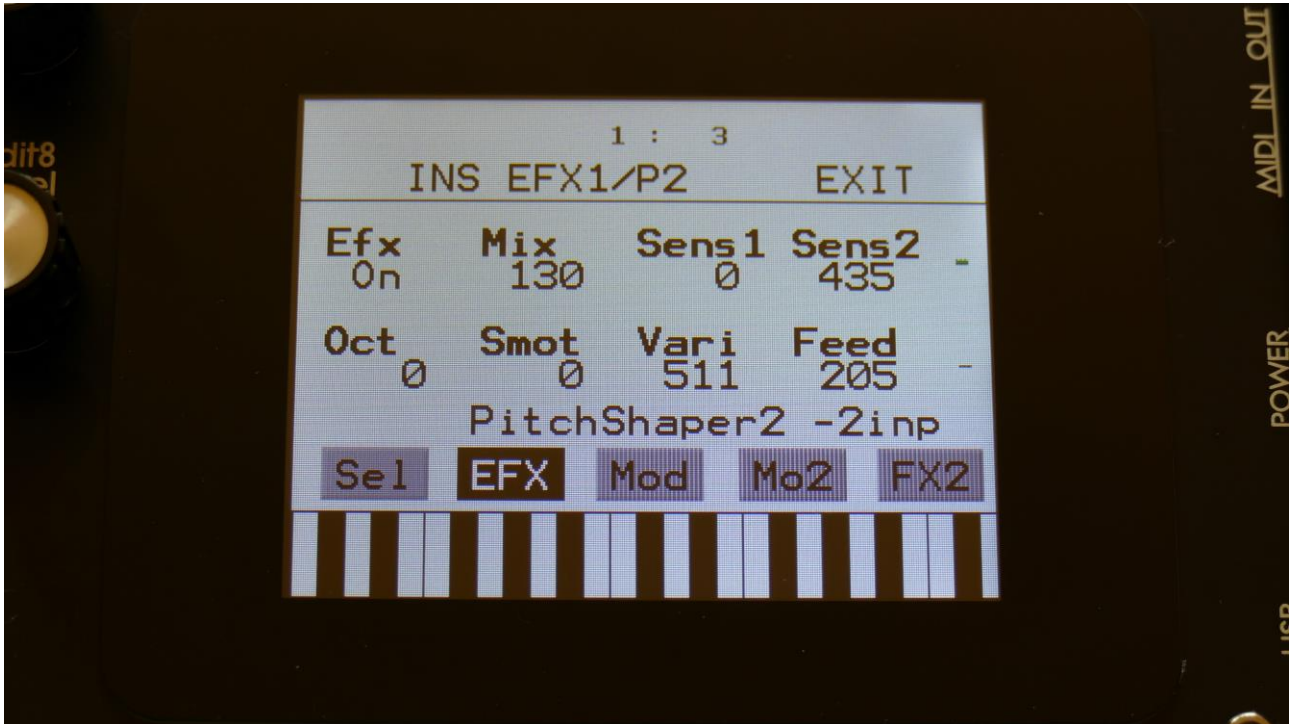
The parameters on this page:

Comp1, Comp2: Will modulate the expansion amount.

Time: Will modulate the expansion “kick in” time.

Rate: Will modulate the time it takes for the expander to react to changes on the input signal.

PitchShaper2



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Sens1: Pitch detection sense input 1. On a pure waveform, turn this fully down to make sure, that it detects all the waves of it, and pitch shifts correctly. On more complex sounds, turn this up until the desired effect are obtained. At higher settings, only portions of the sound will be pitch shifted, and when it doesn't detect any pitch, it will repeat the portion it detected, making the sound "granulate".

Sens2: Pitch detection sense input 2. On a pure waveform, turn this fully down to make sure, that it detects all the waves of it, and pitch shifts correctly. On more complex sounds, turn this up until the desired effect are obtained. At higher settings, only portions of the sound will be pitch shifted, and when it doesn't detect any pitch, it will repeat the portion it detected, making the sound "granulate".

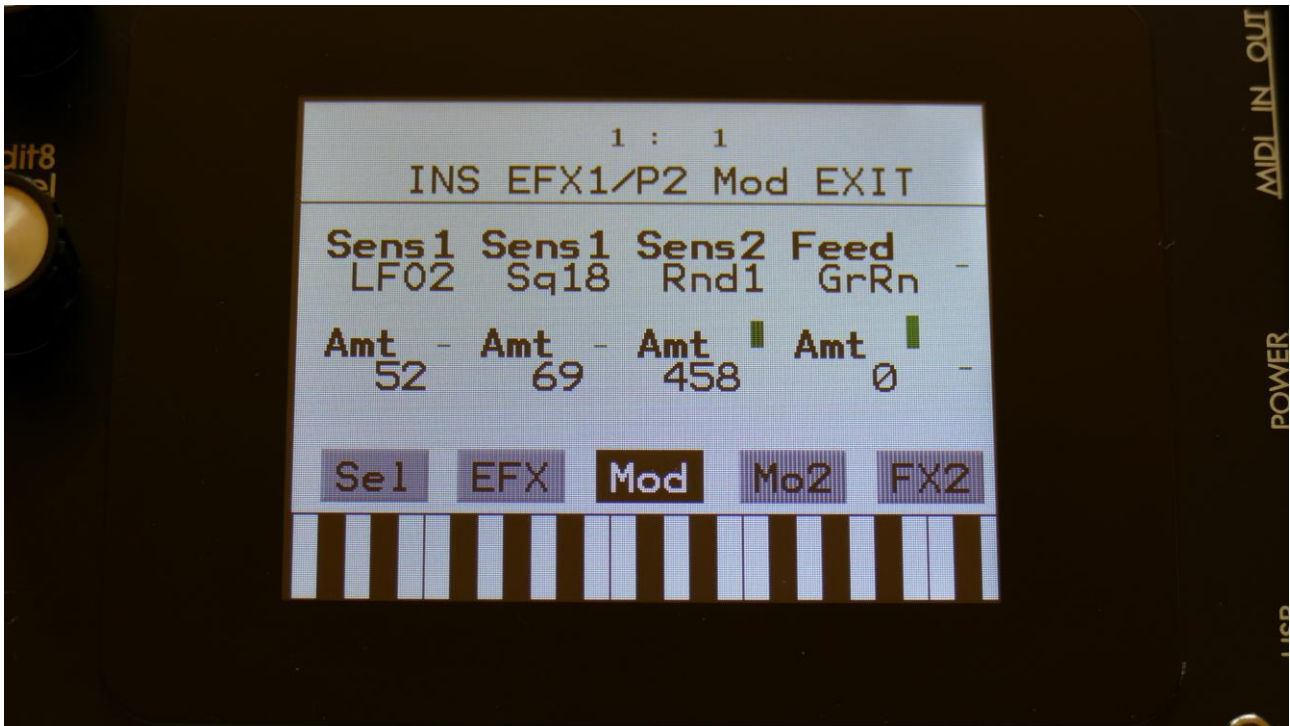
Oct: Octave transpose. From 0 to +3 octaves.

Smot: Smooth. The more this is turned up, the more the changes in pitch are smoothed.

Vari: Pitch variation. The more this is turned up, the more the pitch variations on the input 1 signal affects the pitch shaper frequency.

Feed: Pitch shaper feedback. Adjusts the portion of the output signal, that is fed back to the input.

PitchShaper2 Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

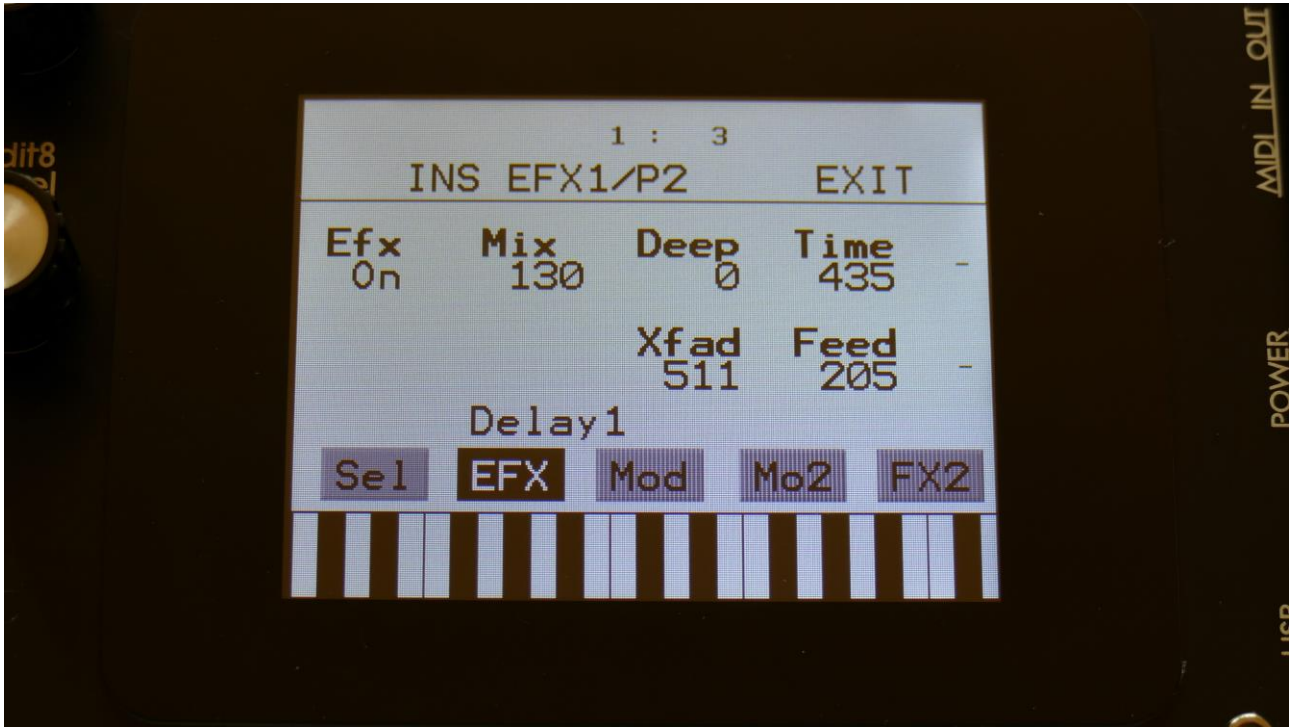
The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Sens1 and Sens2: Modulates the sens 1 and 2 parameters.

Feed: Modulates the Feed parameter.

Delay1, Roto delay, Bright Delay and Bright Roto Delay



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

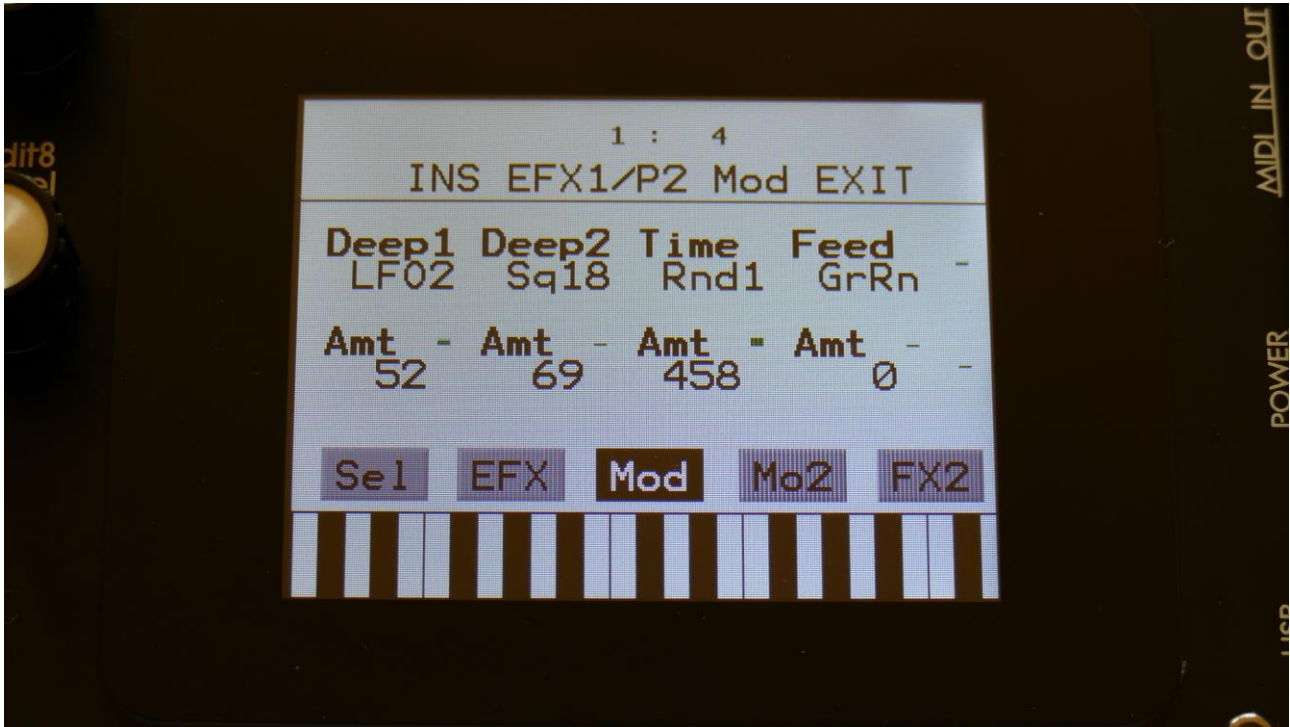
Deep: Simulates a delay box, with adjustable physical depth.

Time: Delay time.

Xfade: Crossfade. Makes the delay effect sound smoother, by crossfading the feedback repeats into each other.

Feed: Delay feedback amount. At values over middle, the feedback signal will be gained and create infinite feedback. This might cause some saturation.

Delay1, Roto delay, Bright Delay and Bright Roto Delay Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

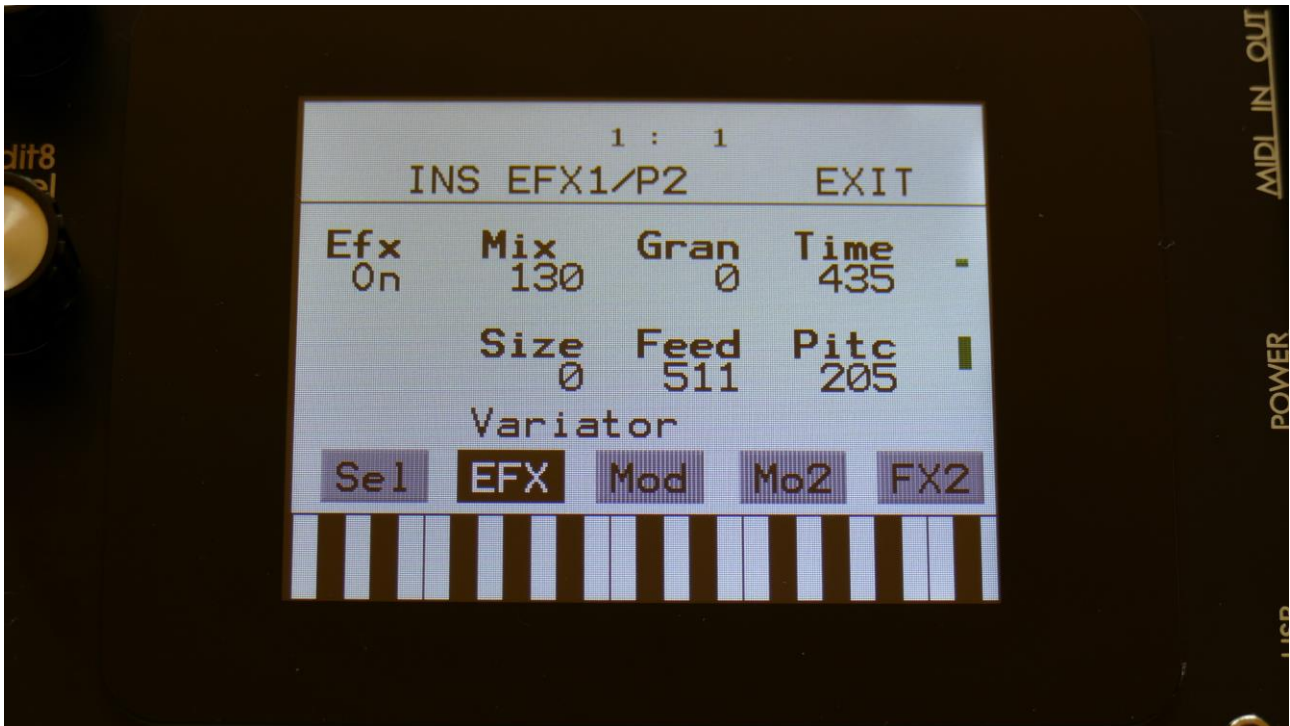
The parameters on this page:

Deep1 and Deep2: Modulates the deep parameter.

Time: Modulates the delay time parameter.

Feed: Modulates the feed parameter.

Variator



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on, Freeze. When the effect is off, it is bypassed, and its input mixer can function as a mixer. When the effect are in Freeze mode, the input signal is no longer sampled. The sound that the effect holds, will just keep playing back on and on again. NOT ALL EFFECTS ARE AFFECTED BY FREEZE!

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output, that is sent to the audio bus.

Gran: Determines the variation pattern. Range: 0 to 511.

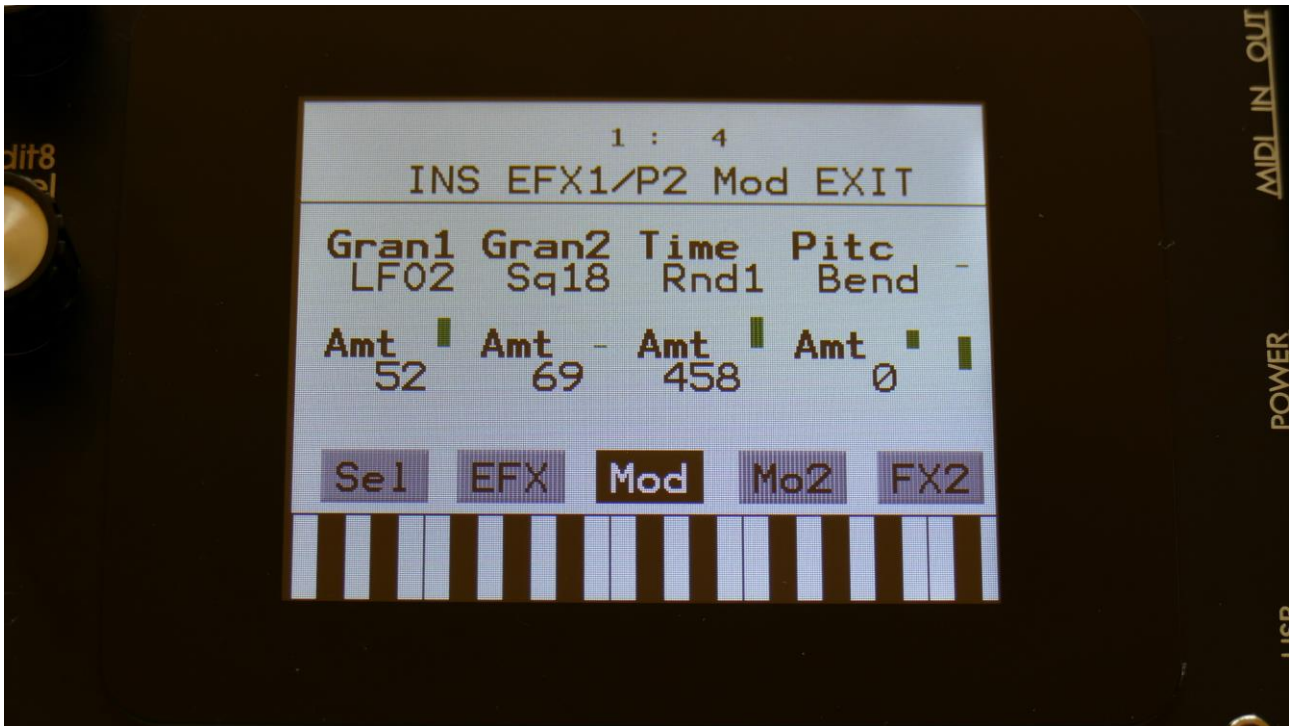
Time: The size of the RAM buffer used for the variator.

Size: The size of one grain.

Feed: Adjusts how much of the Variator output signal should be fed back to its input.

Pitc: The amount of Variator pitch shift.

Variator Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

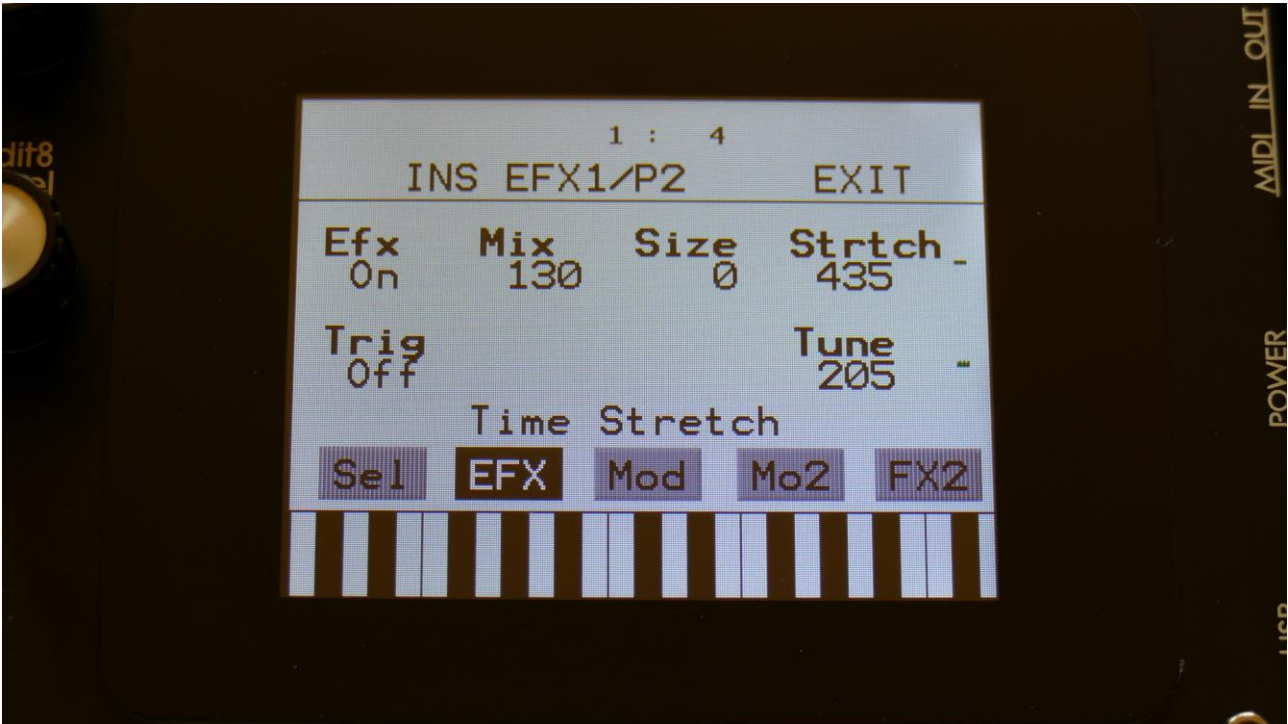
The parameters on this page:

Gran1 and Gran2: Modulates the gran variation pattern parameter.

Time: Modulates the time parameter.

Pitc: Modulates the pitch shift parameter.

Time Stretch



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

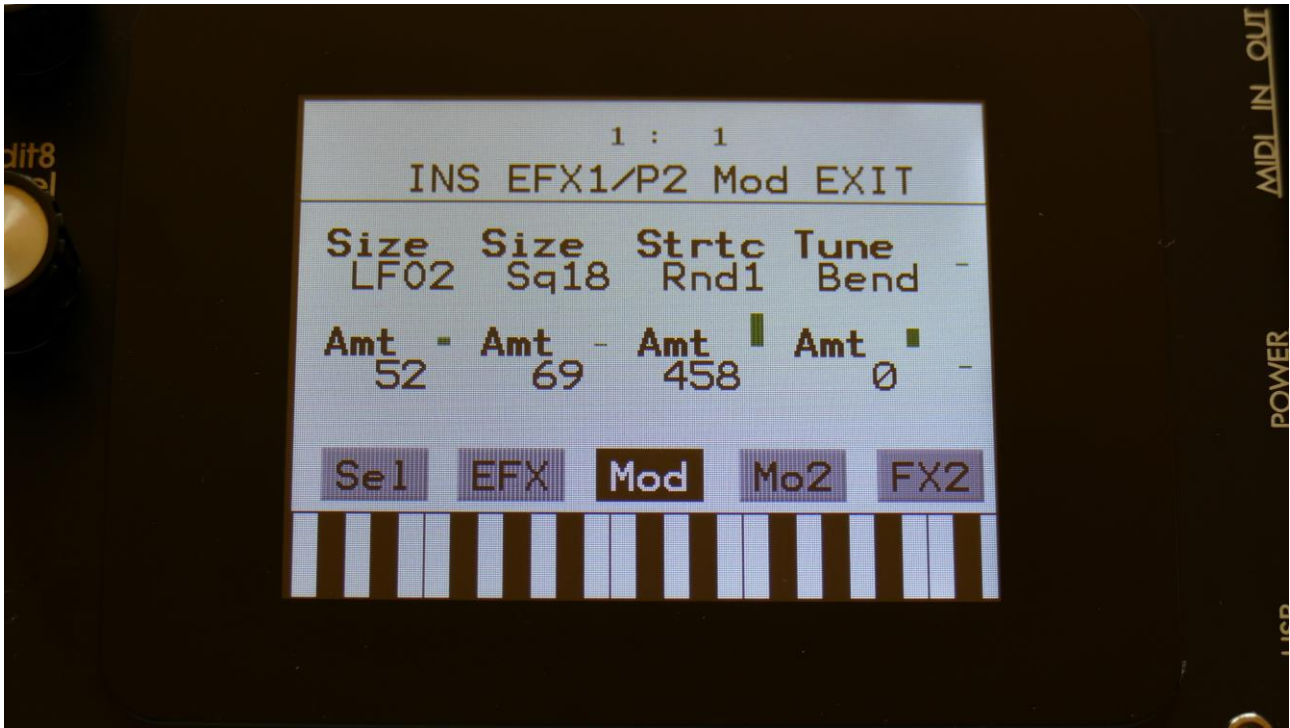
Size: 0 to 511. The size of the RAM buffer used for the TimeStretcher.

Strtch: 0 to 511. Time stretch. A value below 256 will stretch the sound longer, a value above 256 will compress the sound to become shorter.

Trig: Off, 1 to 16. Determines if the time stretch effect should be triggered by a part or not.

Tune: Pitch Coarse. A value below 256 will pitch the signal down, a value above 256 will pitch it up.

Time Stretch Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

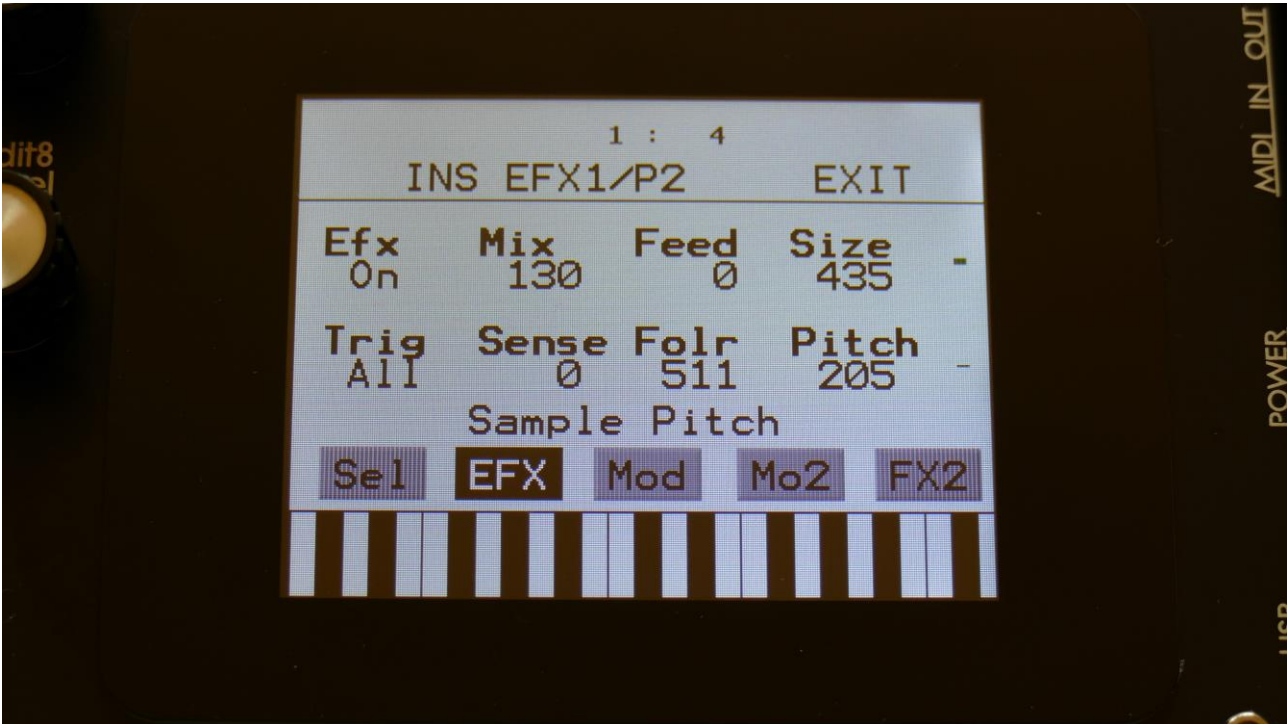
The parameters on this page:

Size: Modulates the size of the time stretch input buffer.

Strtc: Modulates the time stretching.

Tune: Modulates the pitch coarse.

Sample Pitch



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Feed: Sets how much of the effect output signal should be fed back to its input.

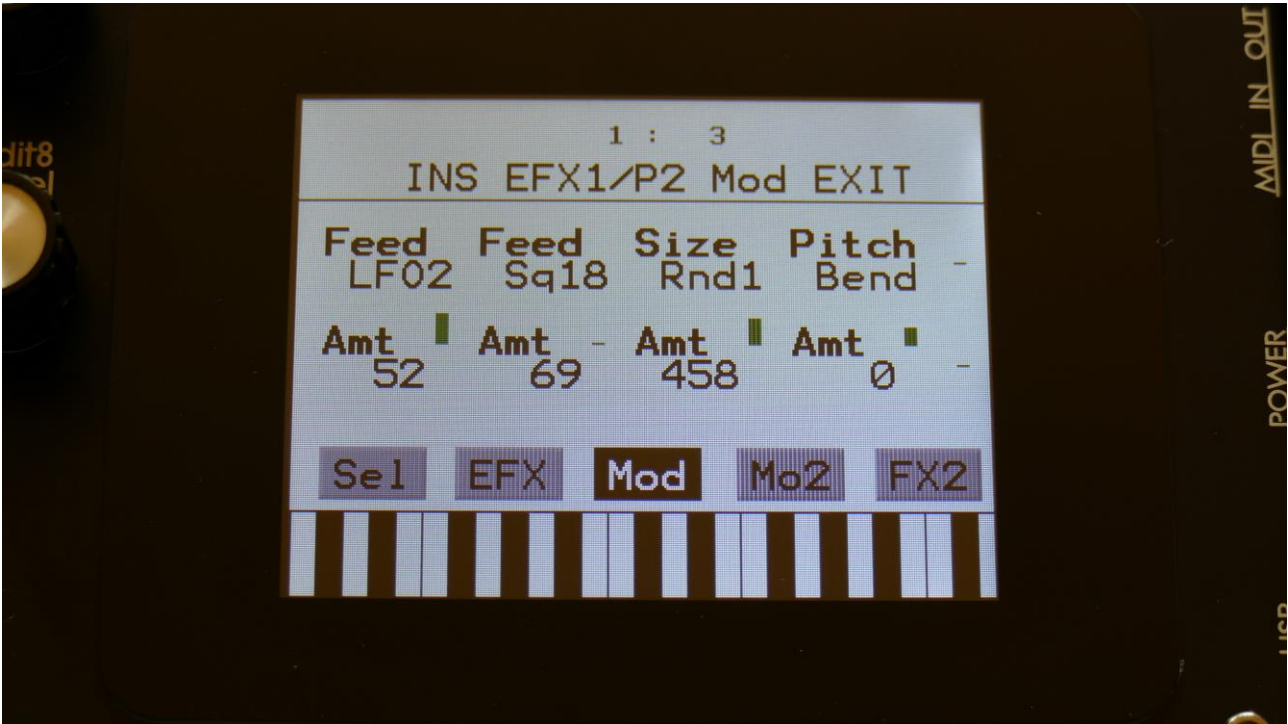
Size: 0 to 511. The number of wave frames used for each pitch grain.

Trig: Off, 1 to 16, OffM, 1M to 16M. Determines if the sample pitch effect should be triggered by all parts or a single part. The "M" suffix indicates that the playback pointer will be moving in another pattern, that will give slightly different results.

Folr: The sample pitch effect output envelope follower rate.

Pitch: The playback pitch. Values below 256 will tune down, values above 256 will tune up.

Sample Pitch Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Feed: Modulates the feed parameter.

Size: Modulates the number of waveframes, used for the pitch shifting.

Pitch: Modulates the pitch shifting.

Output Effects 1 and 2

Spazeboard6 has 2 output effects processors, that part 1 to 8 can be sent to, and that are outputted to the main L and R outputs with pan control.

Both effects has a bypass switch, Freeze modulation, and a mix parameter.

List of Output Effects

Delay 1 – Delay with time and feedback controls, plus Gotharman's Deep, Size, Beam and Xfade controls. Deep adds space to the delay, Size makes the playback range more narrow than the input recording range, and beam beams the delay to previously unknown places. The Xfade control on this delay, creates valleys between the delay taps. The more it is turned up, the more time the valleys takes up.

Roto Delay – This is a 2 tap delay, that is constantly crossfading between the 2 taps. When the Xfade control is turned down, the crossfading is rough, the more it is turned up, the more smooth the crossfading gets. Other controls are the same as the first delay.

Bright Delay – First delay, but with a brighter sound, created by a resonator.

Bright Roto Delay – Roto delay, but with a brighter sound, created by a resonator.

Granulator SQ – Cuts the input signal up in grains, that can be re-arranged. This granulator is synced to the sequencer.

Variator – Creates new variations of the input signal. Both pitch and rhythmic variations.

Reverb – A FAT high resolution synthesized reverb with granular Size parameter.

Granulator (unsynced) - Cuts the input signal up in grains, that can be re-arranged, using the step values of a selectable controller sequencer track.

This is the granulator that was named "AnaGran" in earlier Gotharman instruments. Unlike the Granulator SQ, this is not synced to the sequencer. It plays back one grain, and then it immediately advances to the next grain. The values for the granular sequence, is still taken from a selectable sequencer controller track.

Xfade Granulator - Cuts the input signal up in grains, that can be re-arranged, using the step values of a selectable controller sequencer track. This granulator plays back one grain, and then it crossfades to the next grain.

It is also possible to adjust and modulate the pitch of the granular sequence, using any modulation source.

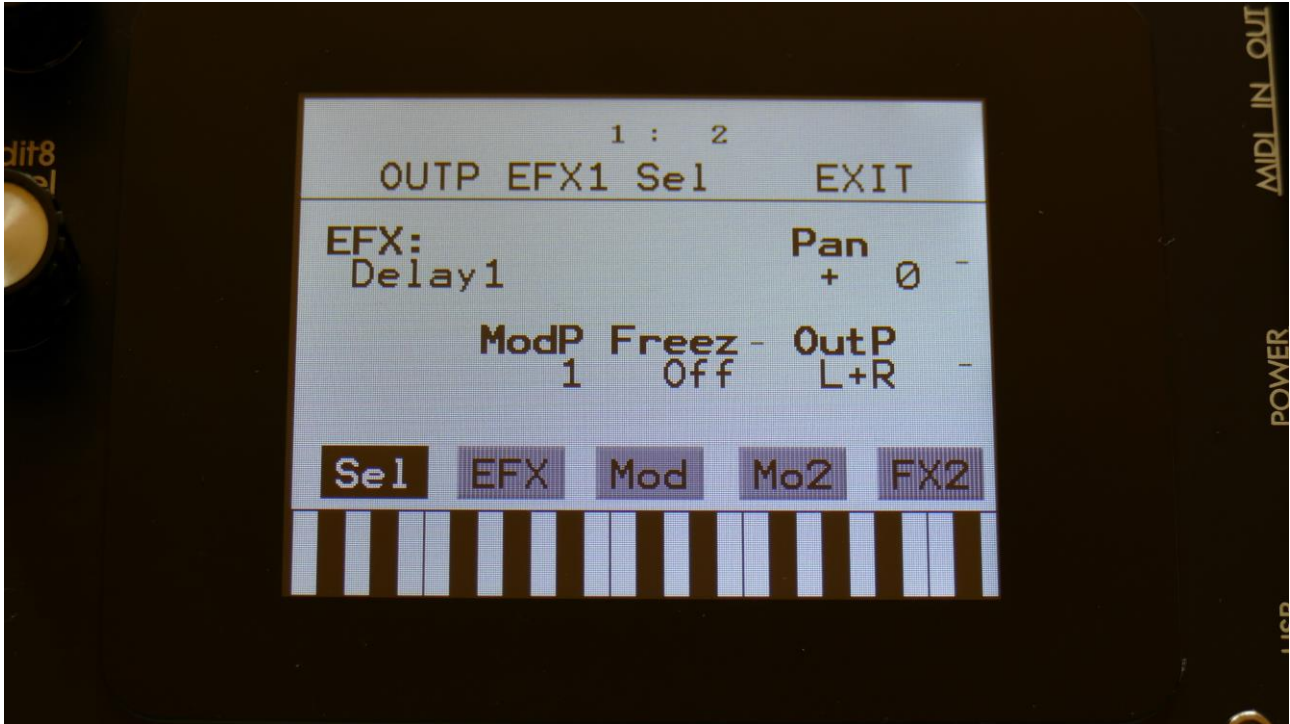
Abstract0 - This effect constantly records the audio applied to its input, in a time interval determined by a "Rec" parameter. It uses 2 "playback heads" to play back the recorded audio. Each of these playback heads can be switched on and off, and the playback length can be adjusted for each of them. Every time a head has played back the adjusted amount of time, it jumps to a new random location of the recorded piece, and starts play back again. The playback pitch of head 2 can be adjusted.

Time Stretch - This effect is able to change the length of the input signal, almost without changing the pitch. Old fashioned time stretch done in realtime, that can be applied even to live input signals! It is also possible to change the pitch if desired, and it is possible to trigger this effect, for bigger precision.

Sample Pitch - Changes the pitch of a sampling, without changing the length, using granular technology. It is possible to set the size of each pitch grain, to generate great granular effects. This effect has to be triggered from any of the parts, in order to function properly. It will work on external signals, but it still needs to be triggered by a part.

Output effects Select page

From the synth parts main page, touch OEFX to enter the output effect pages.



Touch the SEL touch button, to enter the effect select page. Touch the FX2/FX1 button in the lower right corner, to switch between output effect 1 and 2.

The 2 VU-meters at the right of the screen, shows the Effect input and output activity.

EFX: Selects the effect. Please see the List Of Output Effects in the beginning of this section.

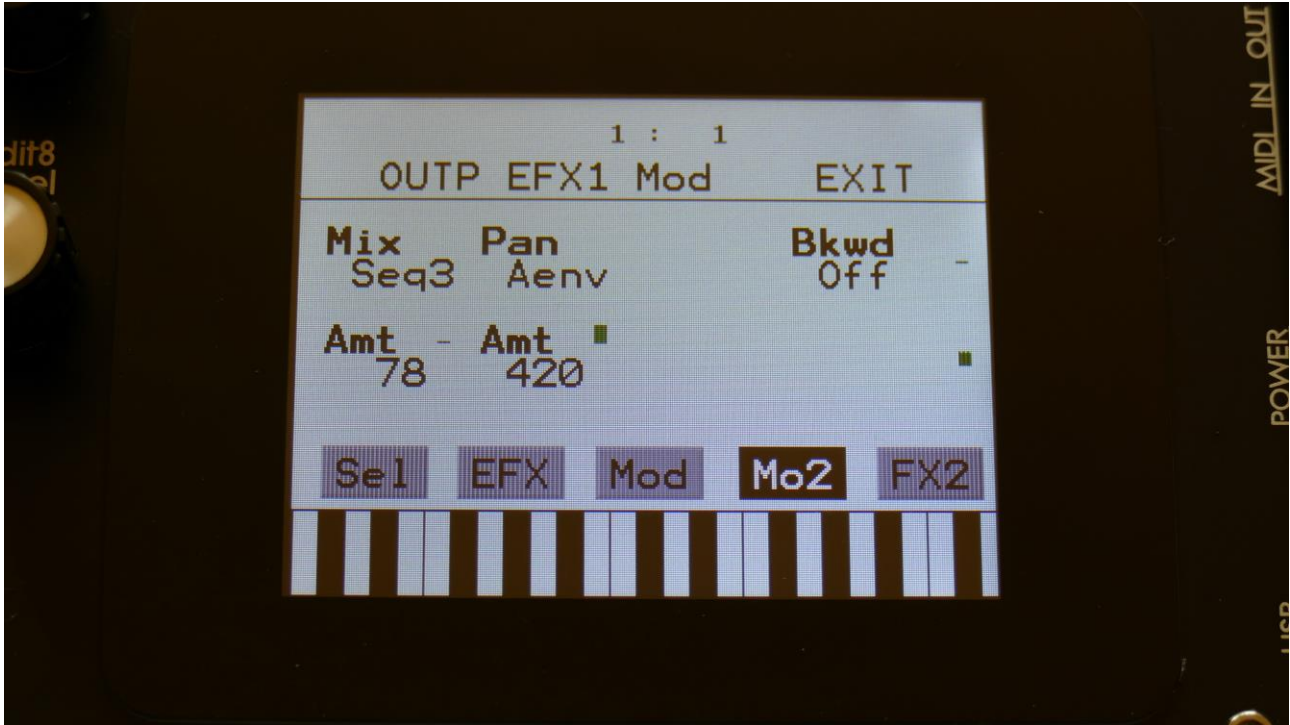
ModP: 1 to 8. Modulation source part. Selects the synth part, from which the output effect will get the modulation sources, selected on the modulation page.

Freez: Freeze modulation. Can be set to off, Freeze button (Buton) or to any modulation source. When set to any other value than off, the effect will freeze, when the modulation source are in its upper range. When an effect freezes, it no longer samples its audio input signal, but only outputs whatever it already holds in its audio buffer.

Outp: It is possible to send the audio output of Output Effect 1 to either the left and right audio outputs, by setting it to **L+R**, or to Output Effect 2, by setting it to **EFX2**, for serial connection of the 2 Output Effects.

Output Effects parameters

Mix and Pan Modulation, Backwards Delay



The Effect dry/wet mix and the panning can be modulated on both Output effects.

Touch the MO2 touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

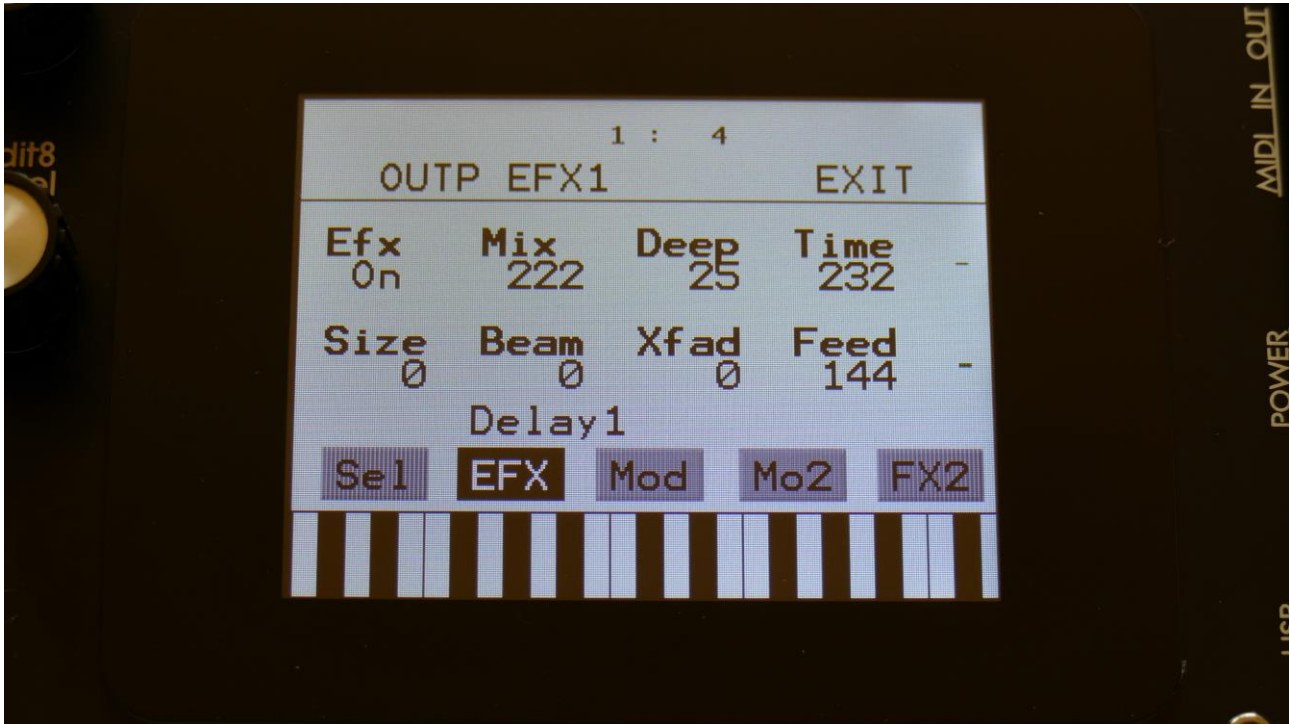
The parameters on this page:

Mix: Modulates the effect dry/wet mix.

Pan: Modulates the effect output panning.

Bkwd: This is only visible, when a delay effect is selected. When turned on, the delay effect will play back backwards.

Delay1, Roto delay, Bright Delay and Bright Roto Delay



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Deep: Simulates a delay box, with adjustable physical depth.

Time: Delay time.

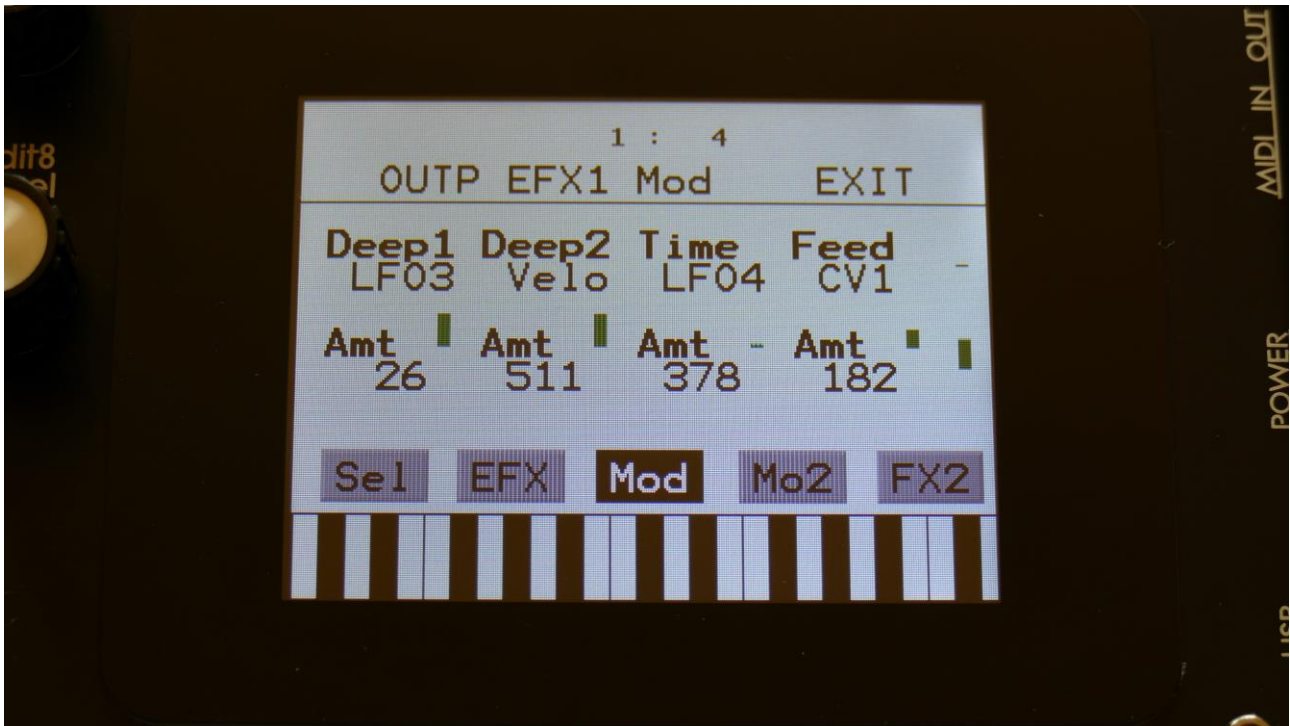
Size: A granular parameter. The more this is turned up, the less space of the delay box is used, causing some echo's to repeat, and others not to sound at all.

Beam: A granular parameter, that "beams" some of the delay sound grains to another place.

Xfade: Crossfade. Makes the delay effect sound smoother, by crossfading the feedback repeats into each other.

Feed: Delay feedback amount. At values over middle, the feedback signal will be gained and create infinite feedback. This might cause some saturation.

Delay1, Roto delay, Bright Delay and Bright Roto Delay Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Deep1 and Deep2: Modulates the deep parameter.

Time: Modulates the delay time parameter.

Feed: Modulates the feed parameter.

Granulator SQ



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Step: Sets how many steps the granulator sequencers should go through, until they starts over again from step one. Range: 1 to 16.

Feed: Adjusts how much of the Granulator output signal should be fed back to its input.

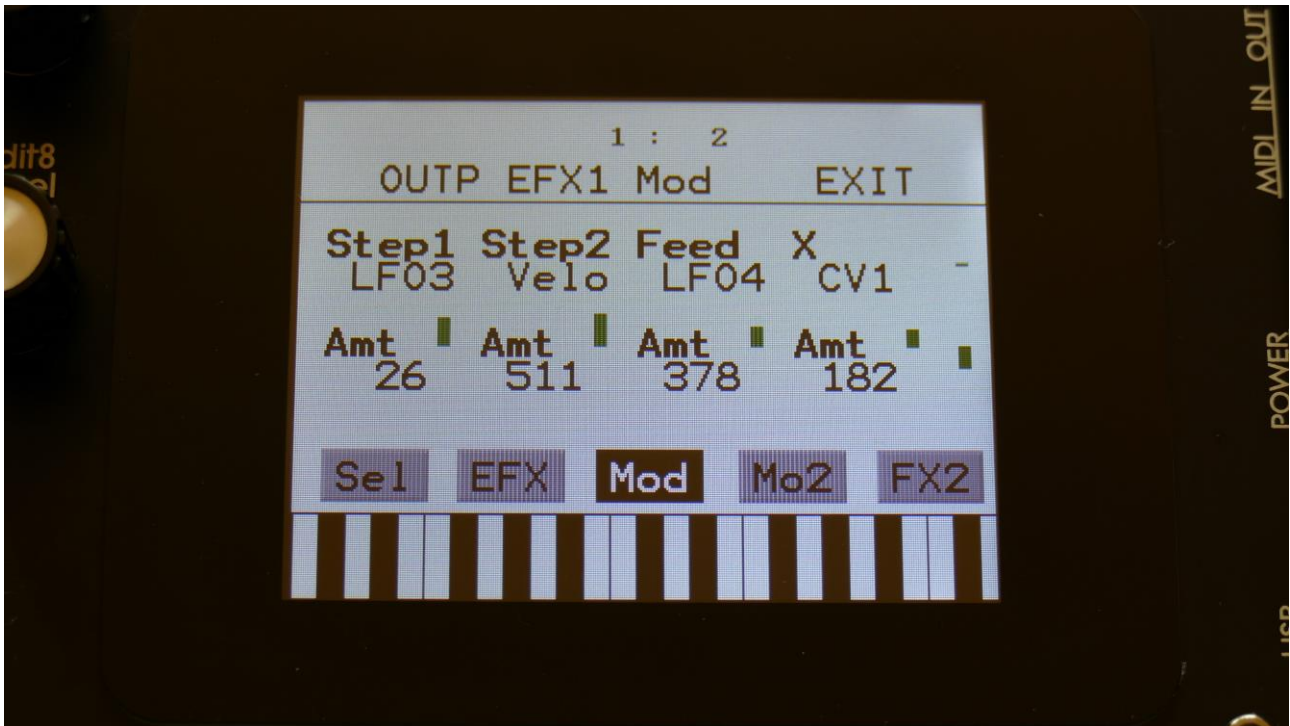
Resolution: Adjusts the resolution of the sequencer track, that controls the granulator. The granulator only uses the controller values of the sequencer steps, so the controller track and the granulator can run in different resolutions.

Size: A granular parameter. The more this is turned up, the less space of each granulator step is used, causing the granulator to “stutter”.

Seq: Selects which of the 16 sequencer controller tracks, the granulator should get its step values from. Each value selects a different portion of the granulator input, to be played back.

X: "X" intermodulation of the sound. Another Gotharman special 😊

Granulator SQ Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

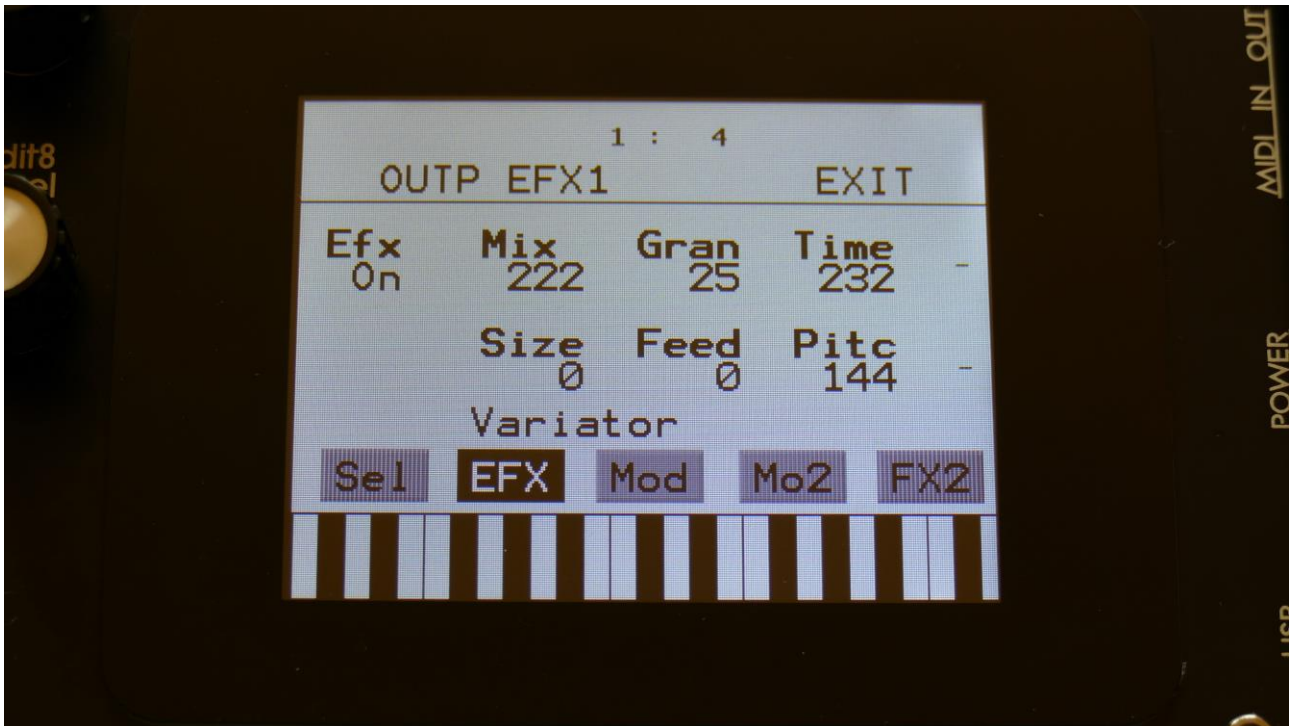
The parameters on this page:

Step1 and Step2: Modulates the number of steps parameter.

Feed: Modulates the feed parameter.

X: Modulates the "X" parameter.

Variator



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on, Freeze. When the effect is off, it is bypassed, and its input mixer can function as a mixer. When the effect are in Freeze mode, the input signal is no longer sampled. The sound that the effect holds, will just keep playing back on and on again. NOT ALL EFFECTS ARE AFFECTED BY FREEZE!

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output, that is sent to the audio bus.

Gran: Determines the variation pattern. Range: 0 to 511.

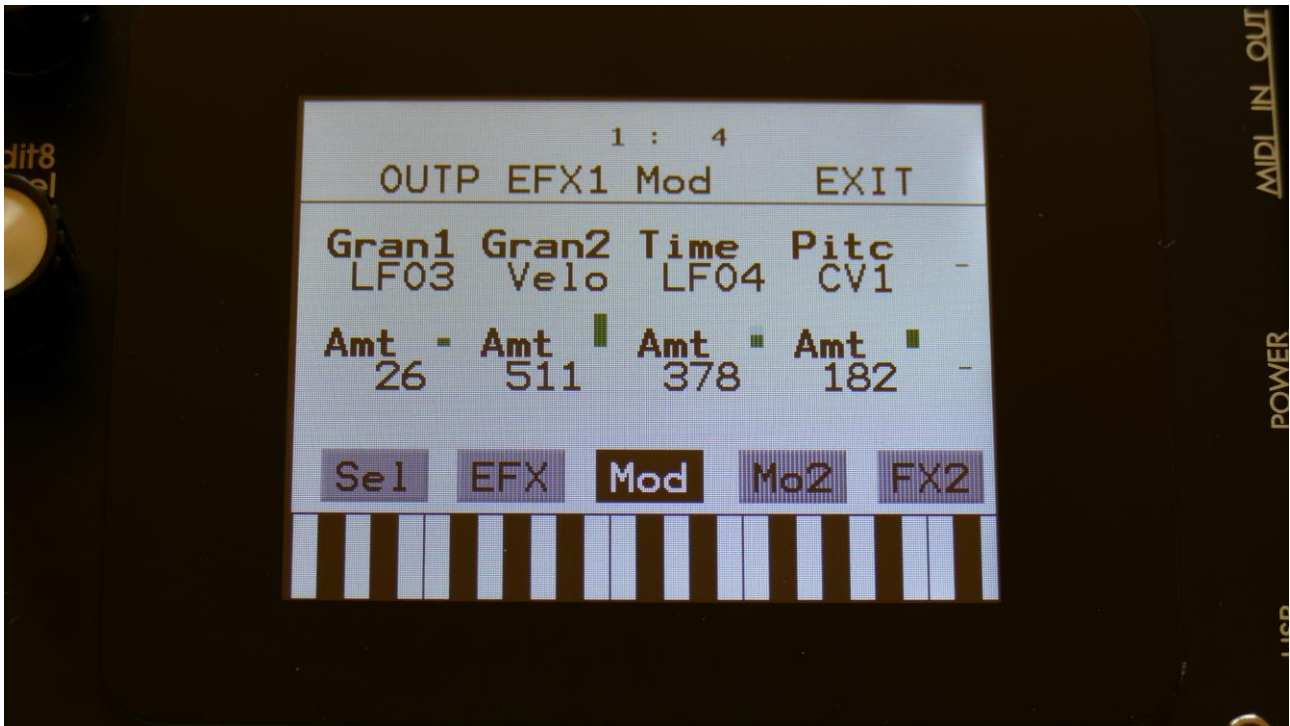
Time: The size of the RAM buffer used for the variator.

Size: The size of one grain.

Feed: Adjusts how much of the Variator output signal should be fed back to its input.

Pitc: The amount of Variator pitch shift.

Variator Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Gran1 and Gran2: Modulates the gran variation pattern parameter.

Time: Modulates the time parameter.

Pite: Modulates the pitch shift parameter.

Reverb



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Feed: Reverb feedback.

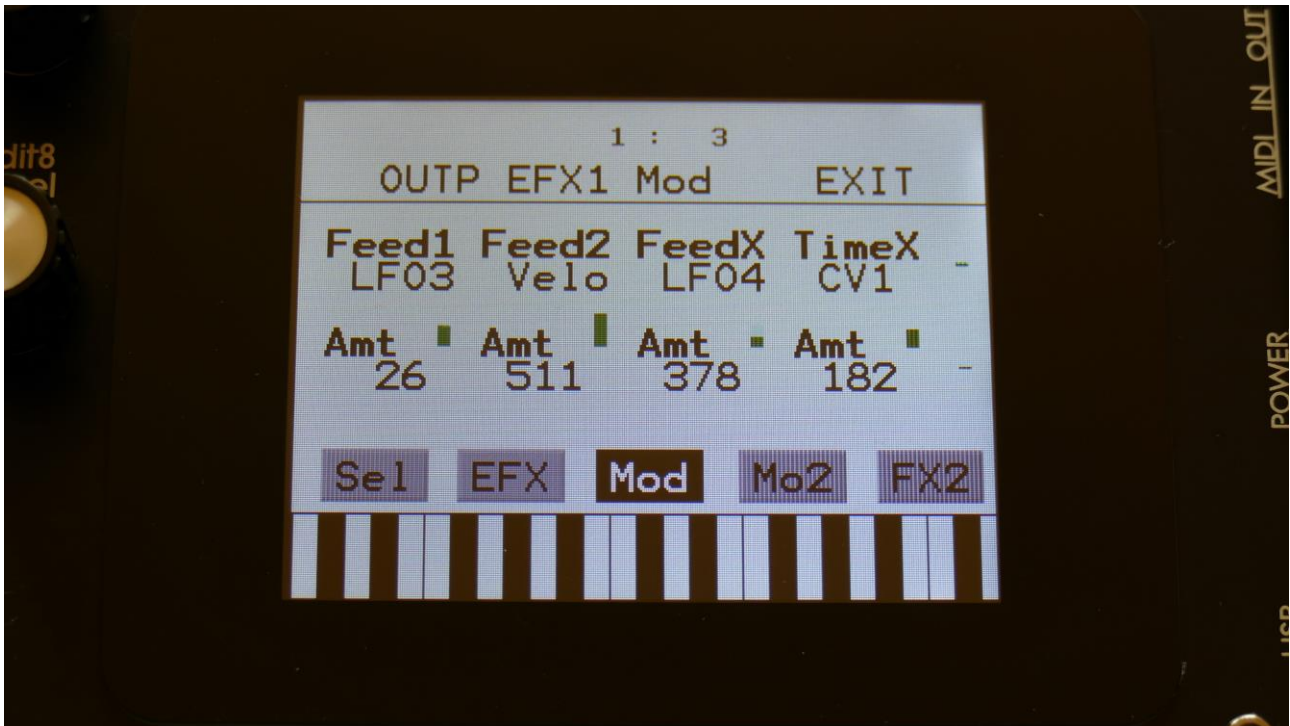
Feed X: Adjusts how much the following reverb taps will decrease in feedback. Lower settings gives a delay effect. As this is turned up, it goes over in a metallic reverb type, and at higher settings it gives a more full reverb.

X: "X" intermodulation of the sound. Another Gotharman special ☺

Time: Reverb time. The total reverb time is a combination of time, timeX, feed and feedX.

Time X: Adjusts how much the following reverb taps will decrease in time. Lower settings is good for a short reverb, and higher settings for a long reverb.

Reverb Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Feed1 and Feed2: Modulates the feedback parameter.

FeedX: Modulates the feed x parameter.

TimeX: Modulates the time x parameter.

Granulator (unsynced)



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Steps: Sets how many steps the granulator sequences should go through, until they starts over again from step one. Range: 1 to 16.

Feed: Adjusts how much of the Granulator output signal should be fed back to its input.

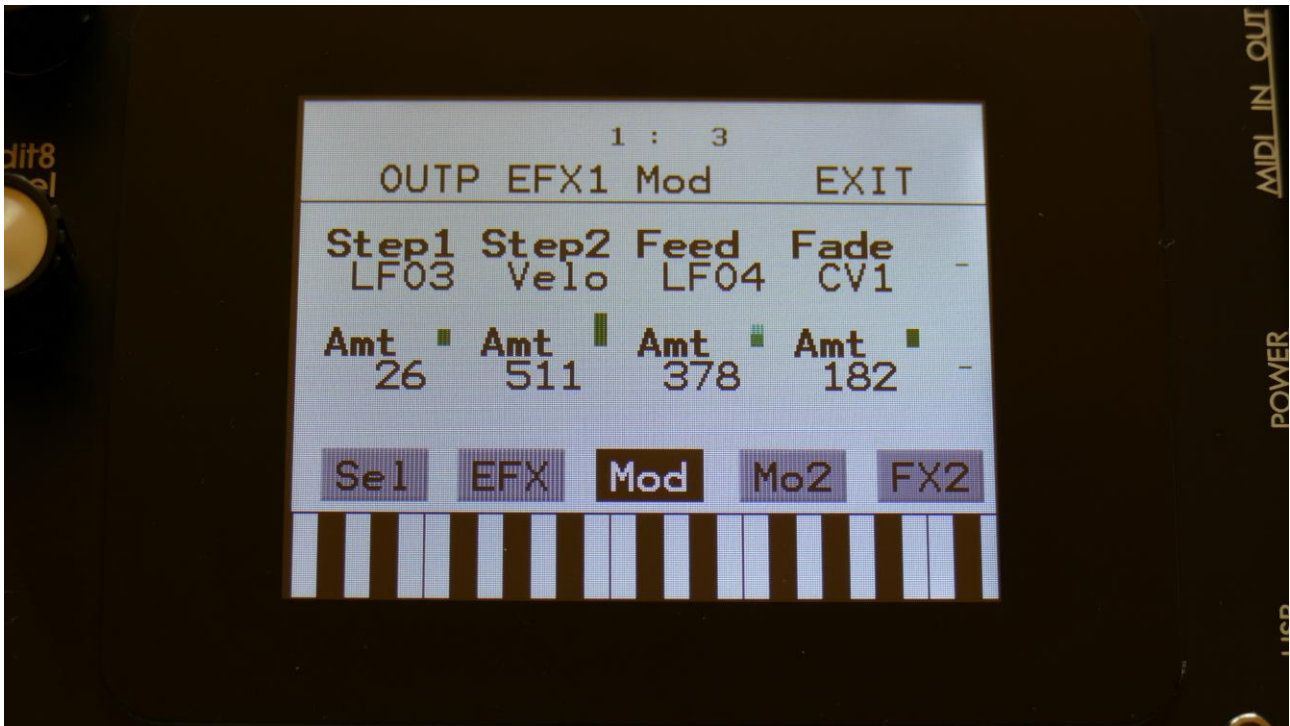
Fragm: Sets how many fragments (or grains) the input signal should be cutted up in. Possible values are 1, 2, 4, 8, 16, 32, 64 or 128.

Time: Sets the size of the input recording buffer. The higher the size, the bigger each grain will be.

Seq: Selects which of the 16 sequencer controller tracks, the granulator should get it step values from. Each value selects a different portion of the granulator input, to be played back.

Fade: When this parameter is turned up, each grain will fade in and out, to minimize clicks.

Granulator Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Step1 and Step2: Modulates the number of steps parameter.

Feed: Modulates the feed parameter.

Fade: Modulates the Fade in and out times.

Xfade Granulator



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Steps: Sets how many steps the granulator sequences should go through, until they starts over again from step one. Range: 1 to 16.

Feed: Adjusts how much of the Granulator output signal should be fed back to its input.

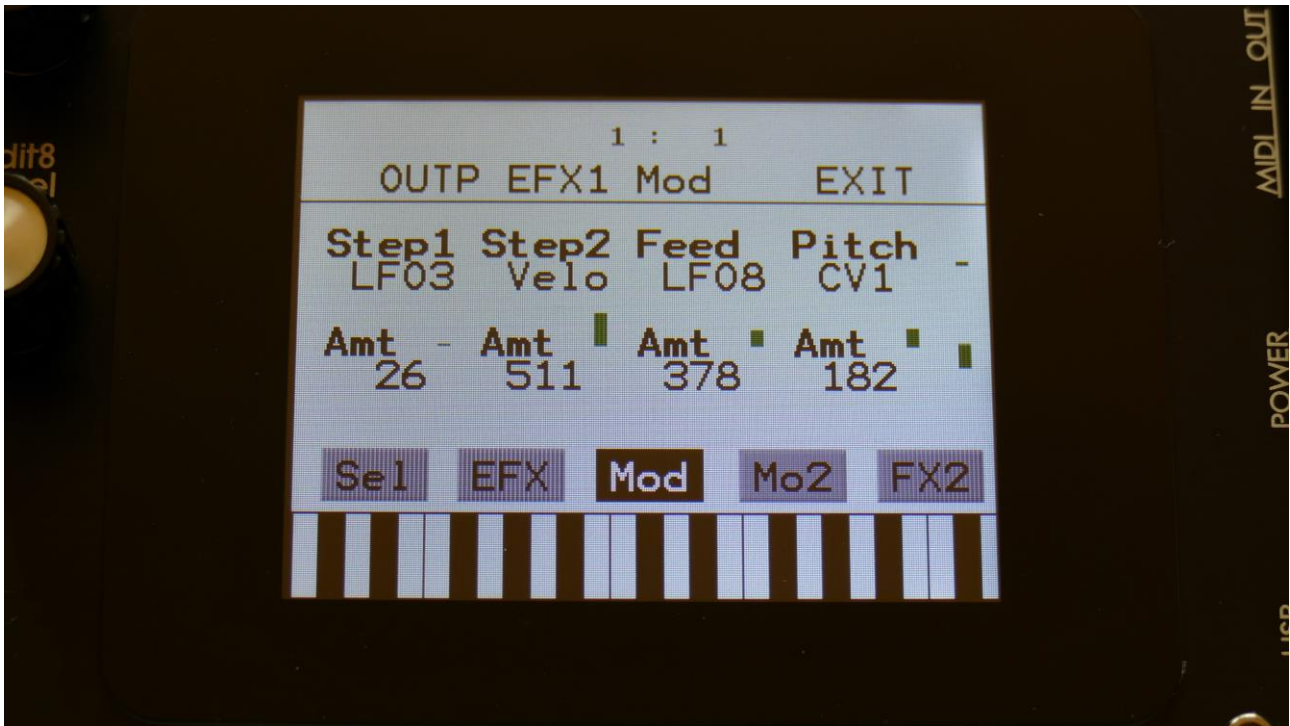
Fragm: Sets how many fragments (or grains) the input signal should be cutted up in. Possible values are 1, 2, 4, 8, 16, 32, 64 or 128.

Time: Sets the size of the input recording buffer. The higher the size, the bigger each grain will be.

Seq: Selects which of the 16 sequencer controller tracks, the granulator should get it step values from. Each value selects a different portion of the granulator input, to be played back.

Pitch: Sets the playback pitch of the granular sequence. A value below 256 will pitch the signal down, a value above 256 will pitch it up.

Xfade Granulator Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Step1 and Step2: Modulates the number of steps parameter.

Feed: Modulates the feed parameter.

Pitch: Modulates the pitch of the granular sequence.

Abstract0



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Play1, Play2: 0 (Off), 1 to 511. Sets for how long time each of the playback heads should play back, until it jumps to a new random location.

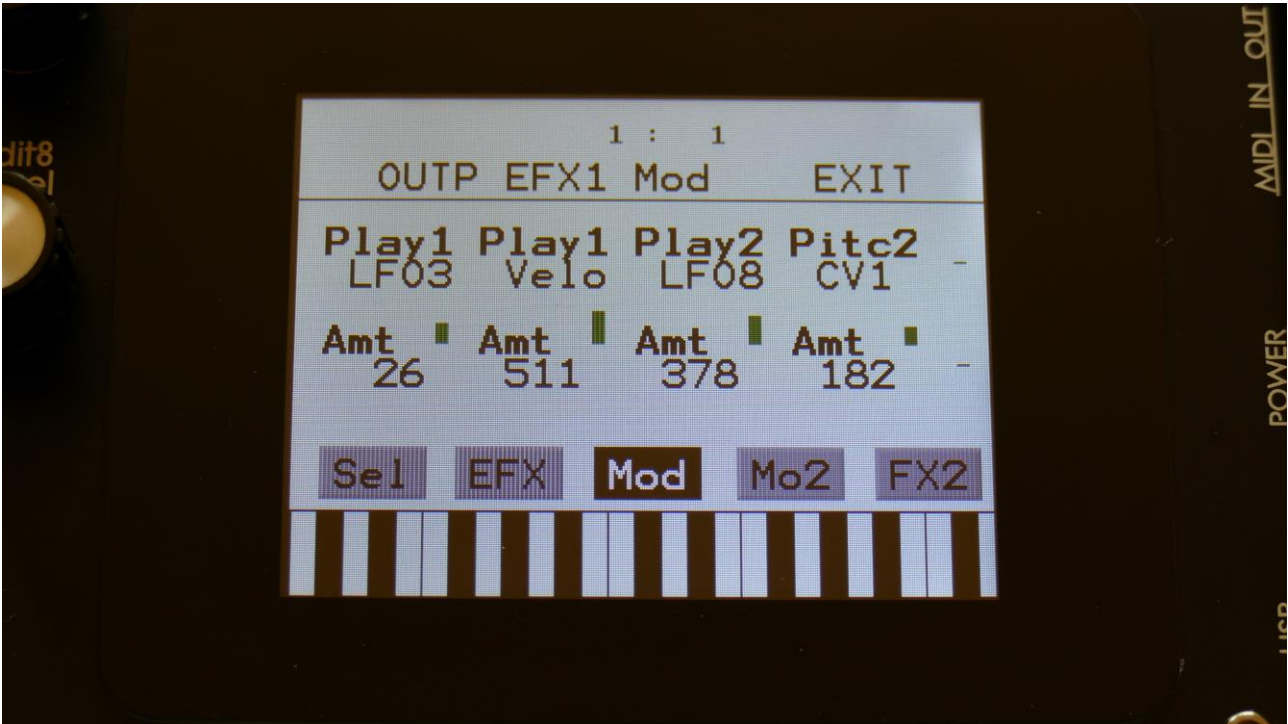
Rec: 0 to 511. Sets the record buffer size. Determines for how long time it should record the input, before it starts over again.

Psize: 0 (Off), 1 to 255. This parameter sets the total playback buffer size, independent of the record buffer size. If this parameter is 0, the Rec parameter determines the playback buffer size, but if it is at any other value, this parameter determines the playback buffer size. It can be used as a fast way of adjusting the playback length of both playback heads at the same time. At low settings, granular effects with very small grains can be obtained. If the value of this is higher than

the Rec parameter, it will play back areas of the record buffer, that is not being recorded, and create a "Freeze" effect.

Pitc2: 0 to 511. Sets the pitch of the playback for playhead 2. A value below 256 will pitch the signal down, a value above 256 will pitch it up.

Abstract0 Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

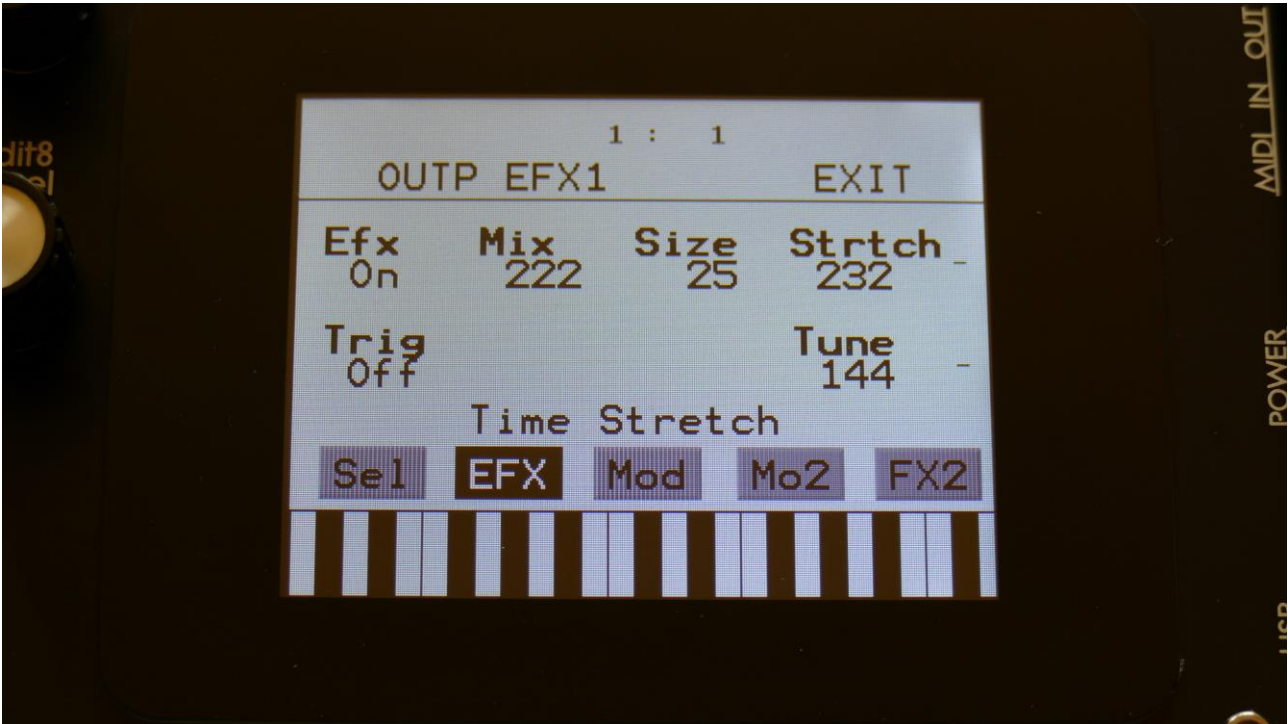
The parameters on this page:

Play1: Modulates the size of playback head 1.

Play2: Modulates the size of playback head 2.

Pitc2: Modulates the pitch of playback head 2.

Time Stretch



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

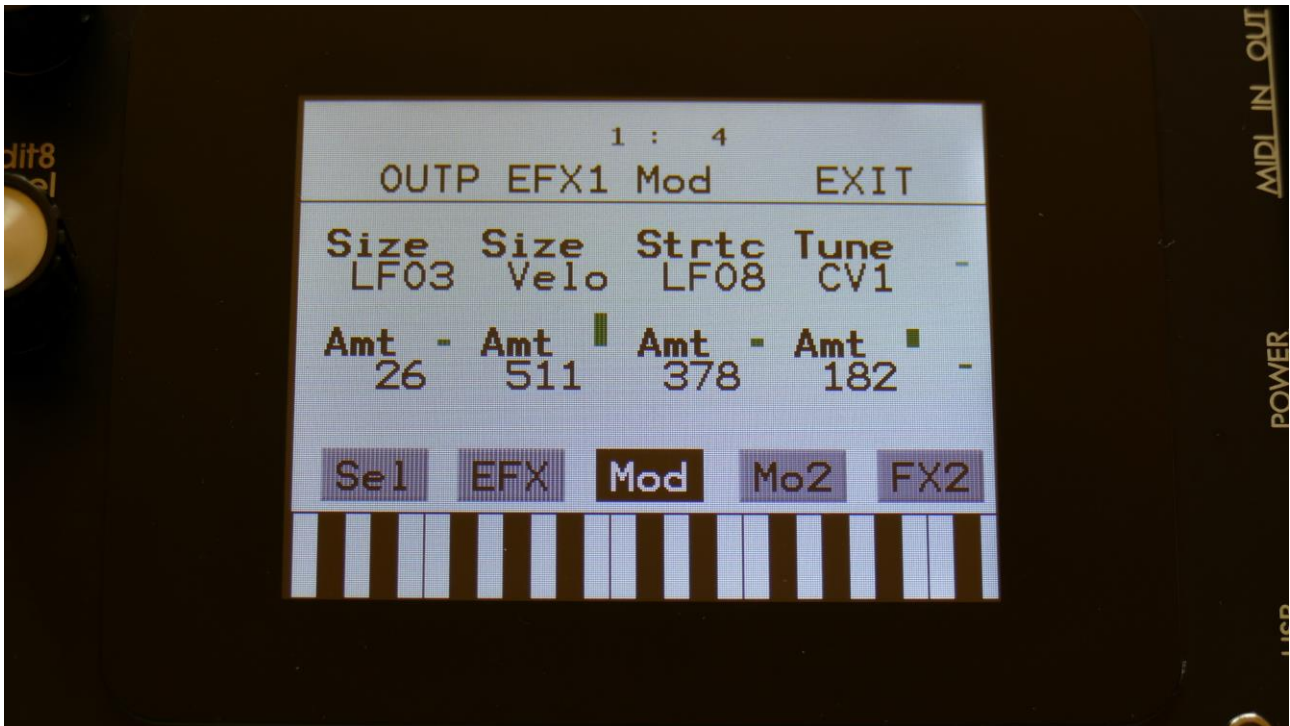
Size: 0 to 511. The size of the RAM buffer used for the TimeStretcher.

Strtch: 0 to 511. Time stretch. A value below 256 will stretch the sound longer, a value above 256 will compress the sound to become shorter.

Trig: Off, 1 to 16. Determines if the time stretch effect should be triggered by a part or not.

Tune: Pitch Coarse. A value below 256 will pitch the signal down, a value above 256 will pitch it up.

Time Stretch Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

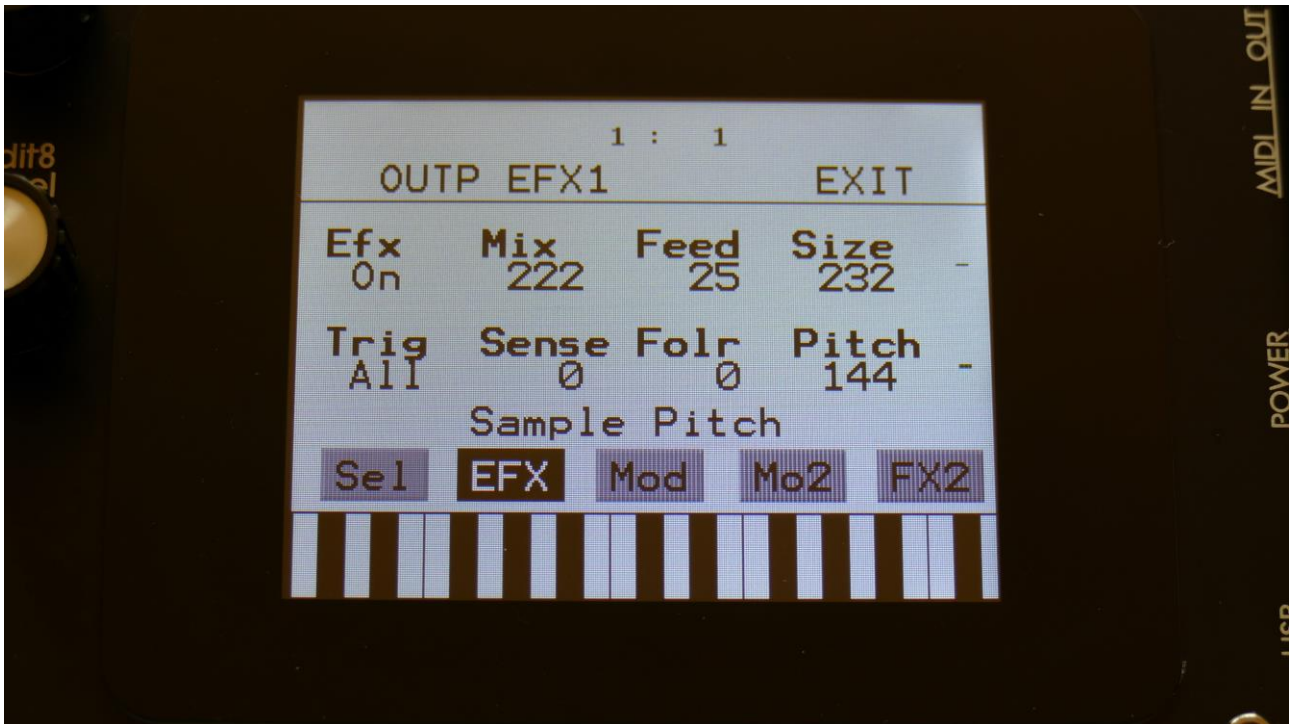
The parameters on this page:

Size: Modulates the size of the time stretch input buffer.

Strtc: Modulates the time stretching.

Tune: Modulates the pitch coarse.

Sample Pitch



Touch the EFX touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output.

Efx: Off, on. When the effect is off, it is bypassed.

Mix: The mix between the un-effected signal on the effect input, and the effected signal on the effect output.

Feed: Sets how much of the effect output signal should be fed back to its input.

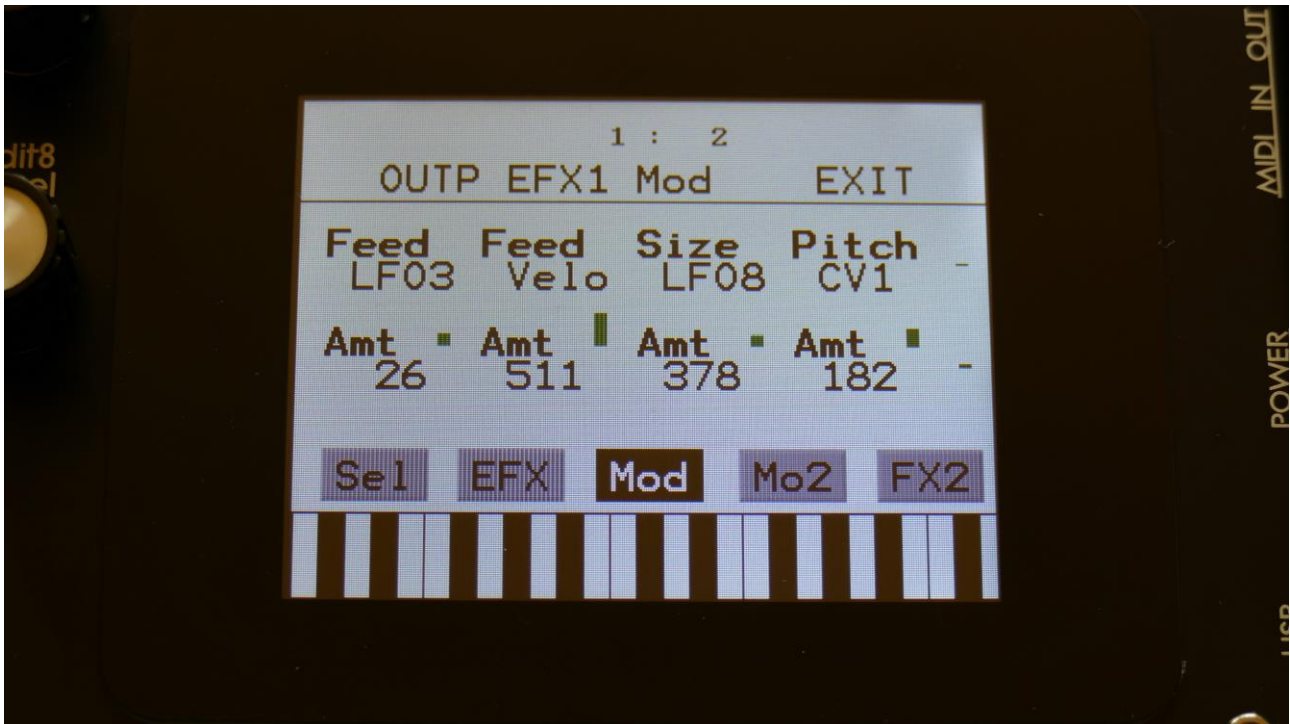
Size: 0 to 511. The number of wave frames used for each pitch grain.

Trig: Off, 1 to 16, OffM, 1M to 16M. Determines if the sample pitch effect should be triggered by all parts or a single part. The “M” suffix indicates that the playback pointer will be moving in another pattern, that will give slightly different results.

Folr: The sample pitch effect output envelope follower rate.

Pitch: The playback pitch. Values below 256 will tune down, values above 256 will tune up.

Sample Pitch Modulation



Touch the Mod touch button, to access this page.

The 2 VU-meters at the right of the screen, shows the Effect input and output. The small VU-meters next to the parameters, shows the activity of the selected modulation sources.

For each parameter, that can be modulated, it is possible to select a modulation source, and to adjust the modulation amount. Only the positive modulation sources, can be selected by the Edit Knobs. To make a modulation source negative, touch the modulation source parameter. For a complete list of modulation sources, see the list in the start of this section.

The upper row of parameters selects the modulation sources, **the lower row of parameters** (Labelled Amt) adjusts the modulation amount in the range 0 to 511.

The parameters on this page:

Feed: Modulates the feed parameter.

Size: Modulates the number of waveframes, used for the pitch shifting.

Pitch: Modulates the pitch shifting.

The Sequencer

The LD3 sequencer has 2 different types of tracks: Note tracks and controller tracks. It has 16 note tracks and 32 controller tracks.

Each note track controls 1 part – Note track 1 controls part 1, note track 2 controls part 2 and so on, and each track has a note number subtrack, a gate time subtrack, a velocity subtrack, a position subtrack, a sub-position subtrack and a step on/off subtrack. Note number, gate time, delay time, swing, velocity and position can be modulated. All 16 note tracks can either control an internal part, or an external MIDI device. Each note track can be up to 64 steps long, and the resolution can be set from 1/64 to 1/2. Each step of a note track, can play back at any position of the sequence, allowing polyphonic step sequencing. A Strum parameter has also been added, to make polyphonic chords strum.

A scale can be added on the output of a note track. This can be a harmonic scale, or a scale that swaps the keys of an octave, or a scale that swaps the black and the white keys.

Each controller track outputs their value to the modulation sources matrix and, if selected, to MIDI out (CC's and PitchBend). They can be up to 128 steps long, and resolution can be set from 1/64 to 1/2. A smooth parameter are available, that when on, will make a controller track slide from one step to the next.

Templates are available, like ramp up, ramp down and randomize. The percentage of how much the template should affect the track is settable.

The position track has special templates for forwards, backwards, alternating, random, 2 note and 4 note sequences.

All tracks has separate last step, start step and resolution settings, so even polyrhythmic sequences are possible.

The tempo of the sequencer can be set between 51 and 306 BPM. It can sync to MIDI clock, and it is both possible to transmit and re-send MIDI clock to MIDI out.

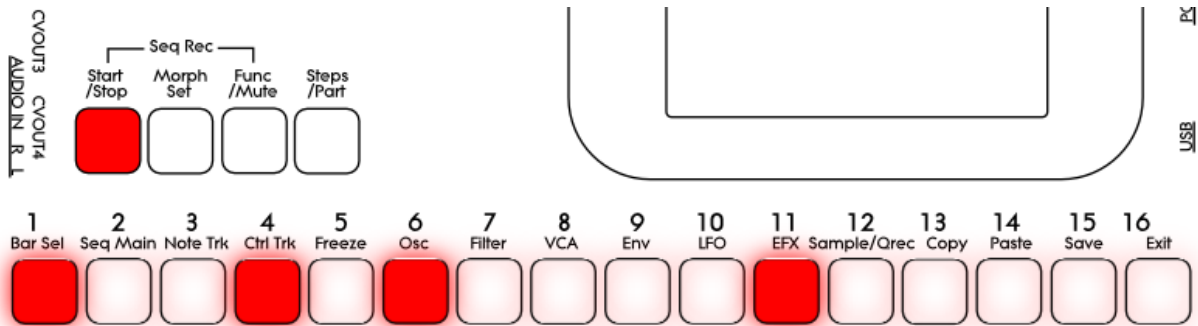
Inputting of note number values can be done in analog/TR fashion, via the step buttons, and it can be done by realtime recording, either monophonic or polyphonic, and by step recording, either monophonic or polyphonic. To assist the realtime recording, a metronome is present. It is also possible to record samplings as audio tracks via the sequencer.

Inputting of controller track values can only be done via the touch screen, by realtime or step recording.

It is possible to mute and edit each track separately.

Remember to save all edits you do in the sequencer. Else they will be lost when you change preset, or turn LD3 off. See how to in the "Save Preset" section.

Starting and stopping the Sequencer:



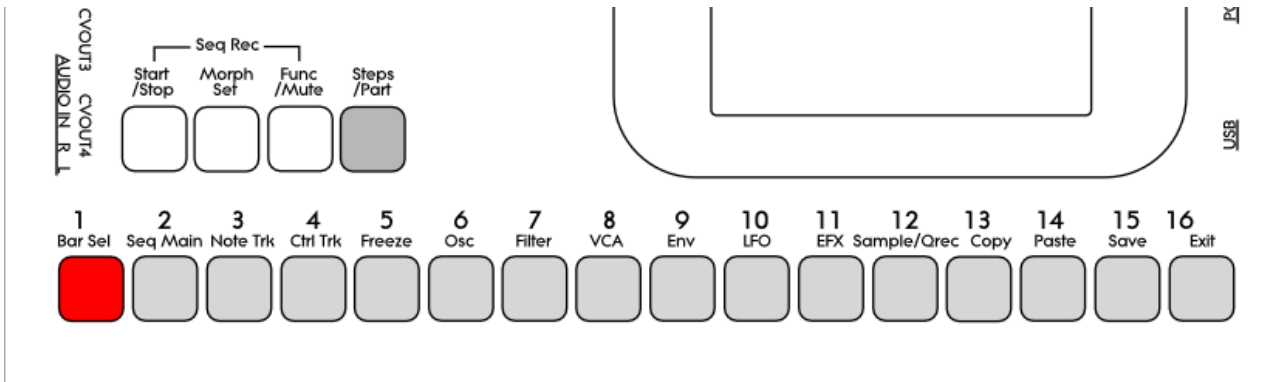
To start the sequencer playback, push and release the Start/Stop button, so that it lights up.

The sequencer will now start to play back. When the Func/Mute and the Steps/part buttons are not lighting up, every time a part is triggered to play back, the corresponding step button will light up shortly.

To stop the sequencer, simply hit the Start/Stop button again, so that it is no longer lighting up.

If you hit the play button to start the sequencer, and it doesn't start, please check in the "MOR>COMMON" section (described later in this manual), that it is not set up to external MIDI sync.

Selecting a track:



Push and hold the Steps/Part button.

The selected track number will now be shown, by one of the 16 step buttons lightning up. The number above the step button, is the track number that is currently selected.

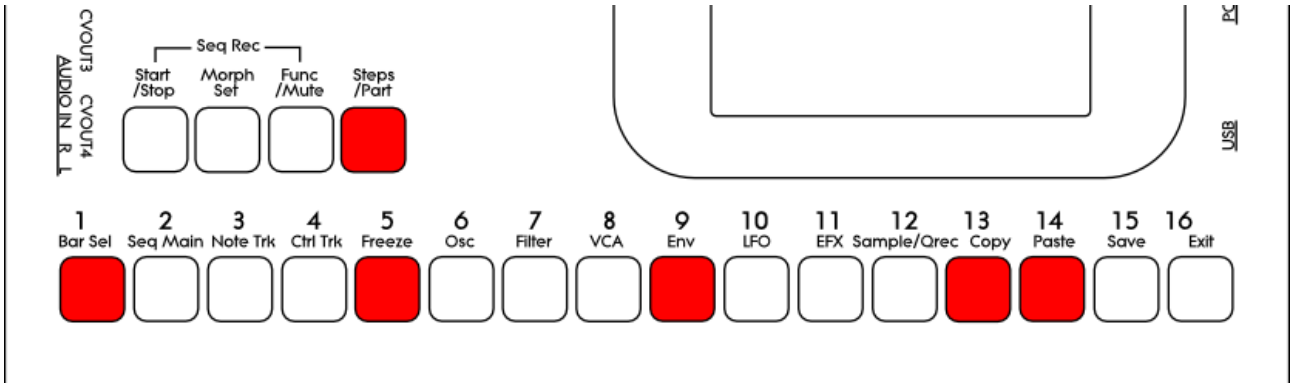
To select another track, while still holding down the Steps/Part button, push any of the 16 step buttons.

The selected track, is the track which parameters will be shown on the display, when entering the edit pages.

When entering the Sequencer Controller tracks 17 to 32, 1 will equal 17, 2 will equal 18, and so on.

Note tracks are selected together with the part, while Controller tracks are selected separately.

Sequencer Note Track Steps view/edit:



First, select the track, for which you would like to view/edit the note steps, as described earlier in this manual. You can, of course, select another track at any time, also after you have entered note step edit mode.

To enter note step edit mode, push and release the Steps/Part button. This should now light up. Any note sequencer steps that are switched on to play back, will now also light up on the 16 step buttons.

If the sequencer is running, the light state of each step button will be reversed, when a step is playing back.

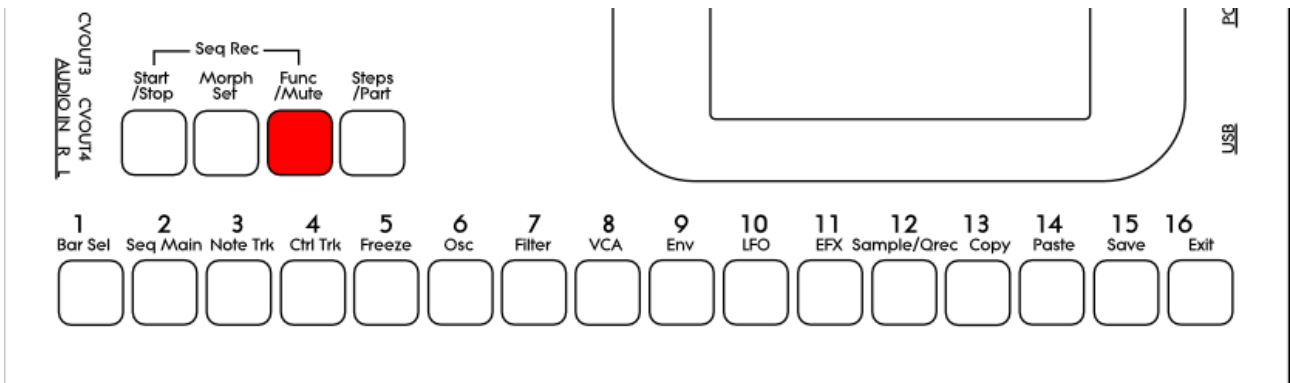
To switch a step on or off, simply hit the corresponding step button, at it will toggle its state.

When the sequencer are in recording mode (the Start/Stop button is flashing), pushing any of the step buttons, will set record mode to step mode, and select this step for recording. The step button will now flash.

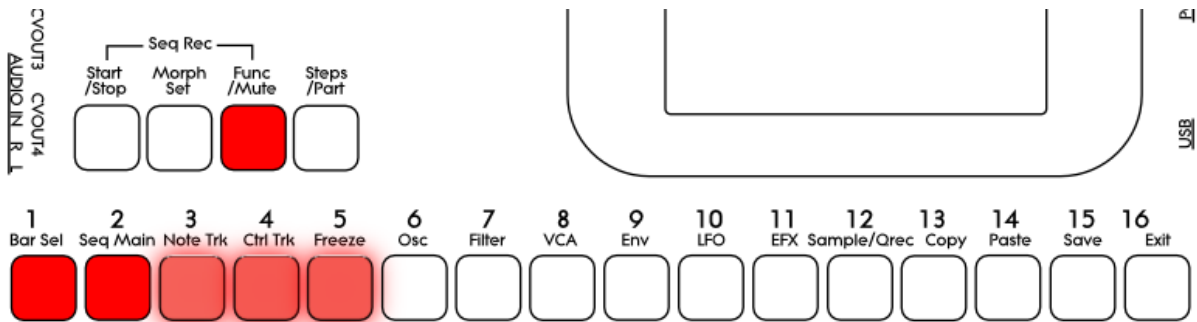
When using the LD3 sequencer as a usual step sequencer, the steps will play back from left to right. In this case, step 1 plays back at position 1, step 2 plays back at position 2 and so on. On the LD3 sequencer, it is though possible to break this pattern, and make each step play back on any position, using the position subtrack. It is even possible to make more steps playing back at the same position, for polyphonic step sequencing.

Only 16 steps are shown at a time. The note tracks of LD3 has 64 steps. Please read on, to experience how to switch between step 1-16, 17-32, 33-48 and 49-64.

Selecting Sequencer Note Track steps 1-16, 17-32, 33-48 and 49 to 64 / + Controller tracks steps 65-80, 81-96, 97-112, 113-128:



Push and release the **Func/Mute** button, so that it is lightning up. Now, push and release step button 1, **Bar Sel**:



Now step button 1, **Bar Sel**, will light up.

Note tracks: When you are located on any other pages, than the sequencer controller tracks pages, bar is selected to be viewed for a note track. The selected bar, is the bar that is shown on the step buttons, and on the note track pages.

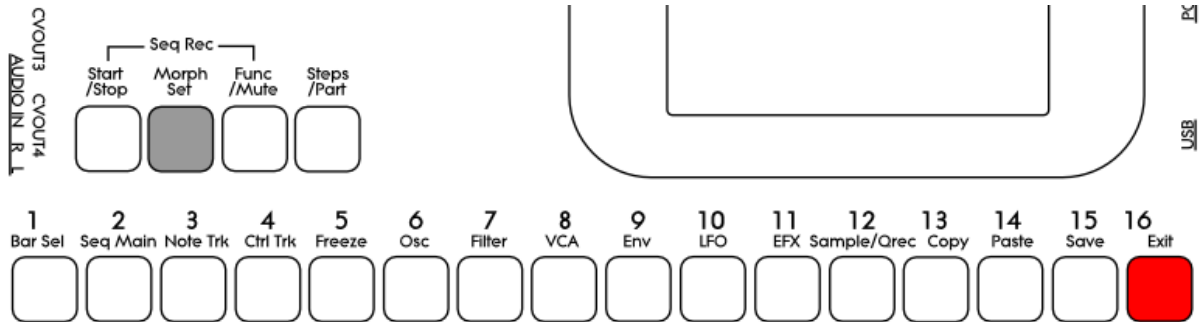
Hit Bar Sel, so that it lights up. Now step buttons 2 to 5 will select bar 1 to 4, of the currently selected track. The button of the bar that is selected, will light up. The other 3 bar select buttons will flash.

Controller tracks: When you are located on any of the controller track pages, bar is selected to be viewed for the currently selected controller track. The selected bar, is the bar that is shown on the controller track pages.

Hit Bar Sel, so that it lights up. Now step buttons 2 to 9 will select bar 1 to 8, of the currently selected track. The button of the bar that is selected, will light up. The other 7 bar select buttons will flash.

When pushing a bar button, that is already selected, both on note and controller tracks, the function of the Edit Knobs will now be toggled: Parameters > Knobs1-8 > Knobs 9-16.

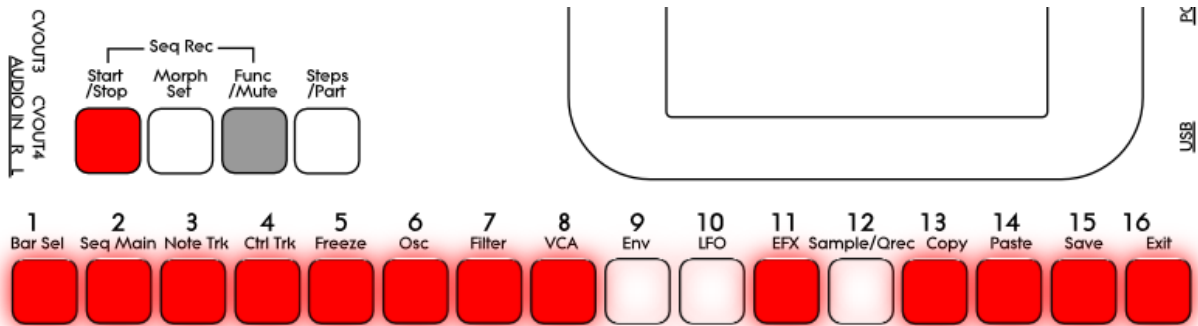
Setting the last step of a Note Track



This can be done on the Note Tracks edit pages, but it can also be done via the pushbuttons.

Push and hold the Morph Set button. If the last step of the selected note track are inside the selected step range, this will now be shown, as a step button that lights up. Push any step button, while still holding down the Morph Set button, to set the last step of the selected Note Track.

Mute/Unmute Note Tracks:



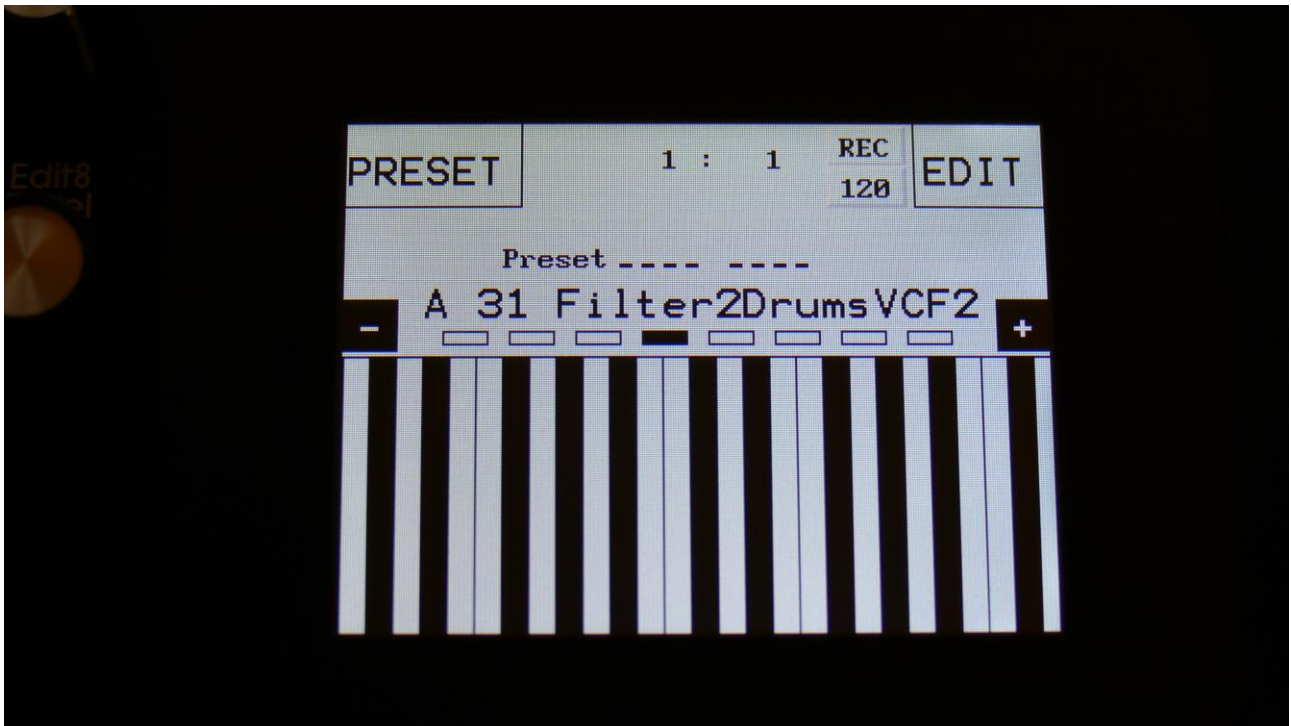
To mute, unmute or view the mute state of the 16 note tracks, push and hold the Func/Mute button.

Unmuted tracks will now be shown by a step button that is lighting up, and shortly flashes off, every time the track is triggering.

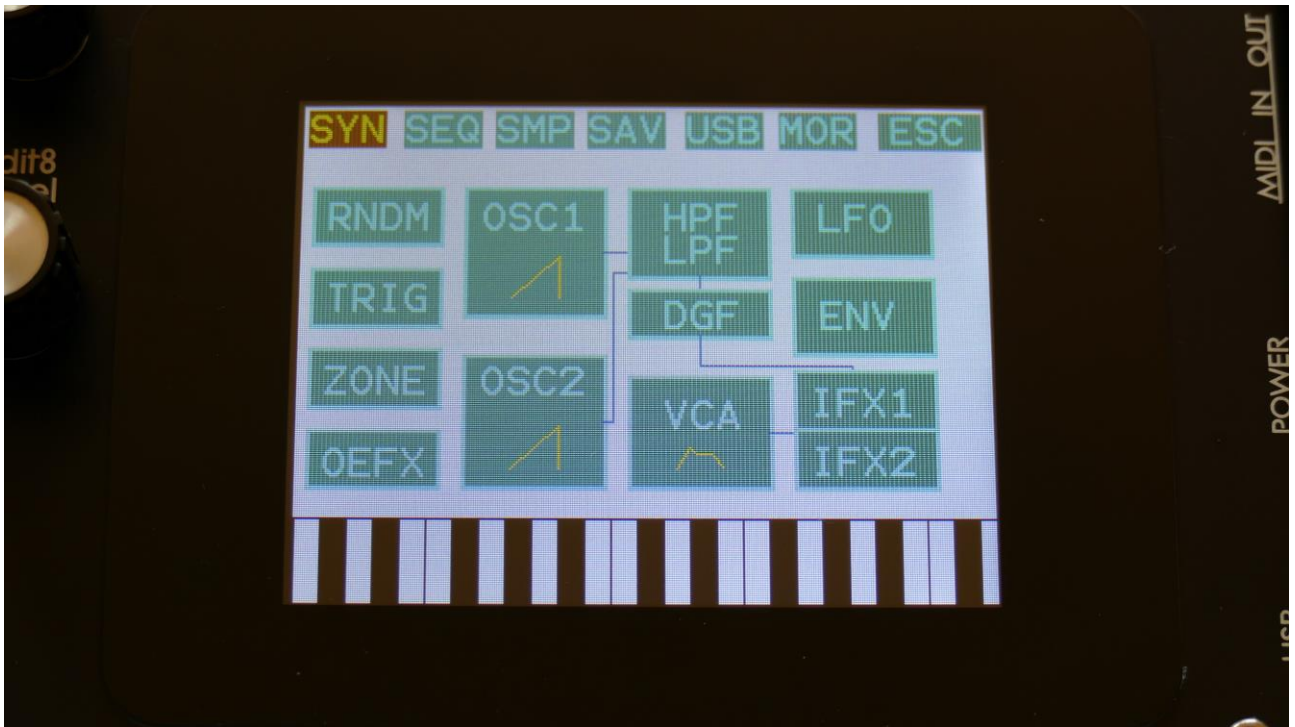
Muted tracks are shown by a step button that is unlit, and that lights up shortly, every time the track would have triggered something, if it weren't muted.

To mute or unmute a track, simply hit the corresponding step button, while still holding down the Func/Mute button.

Entering the Sequencer

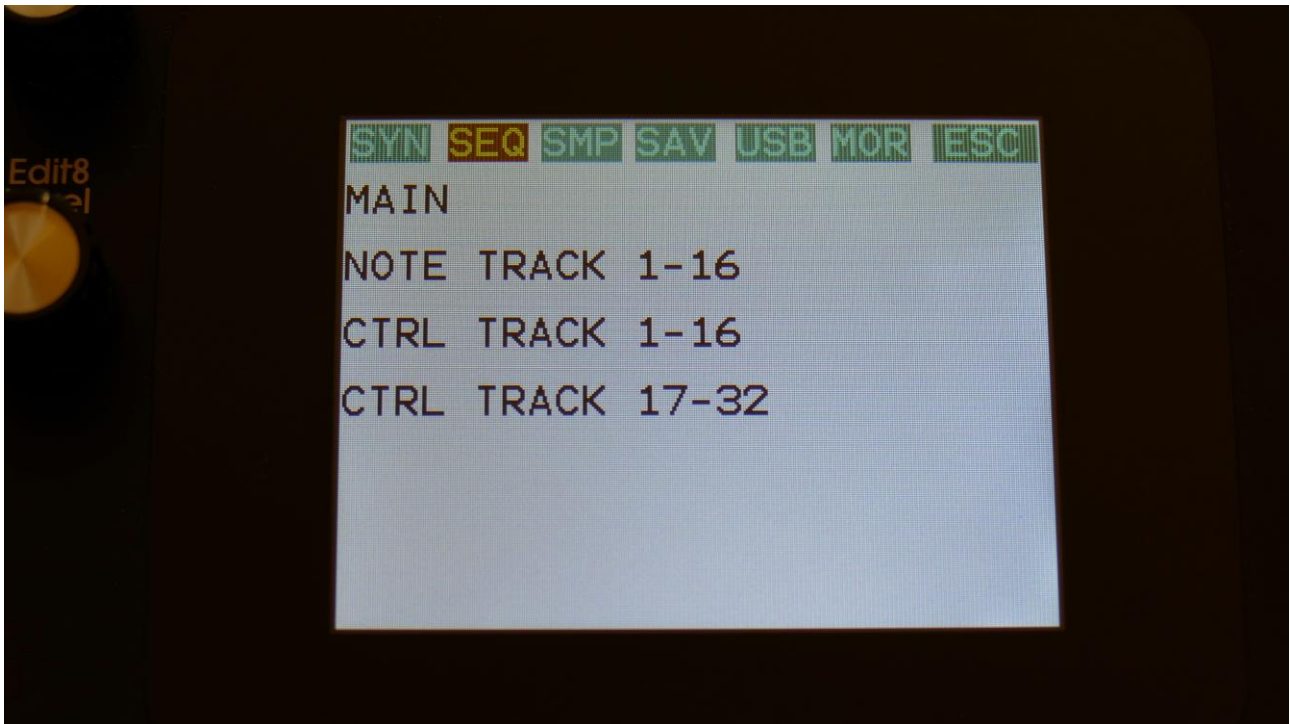


From the Preset Select Screen, touch the EDIT field.



Touch SEQ on the navigation bar in the top of the screen.

This will bring you to the Sequencer access page:



SEQ on the navigation bar has now changed its colour to brown/yellow, indicating that you have entered the Sequencer pages.

To enter the Sequencer Main page or any of the tracks, simply touch the text showing the name of the group that you would like to enter.

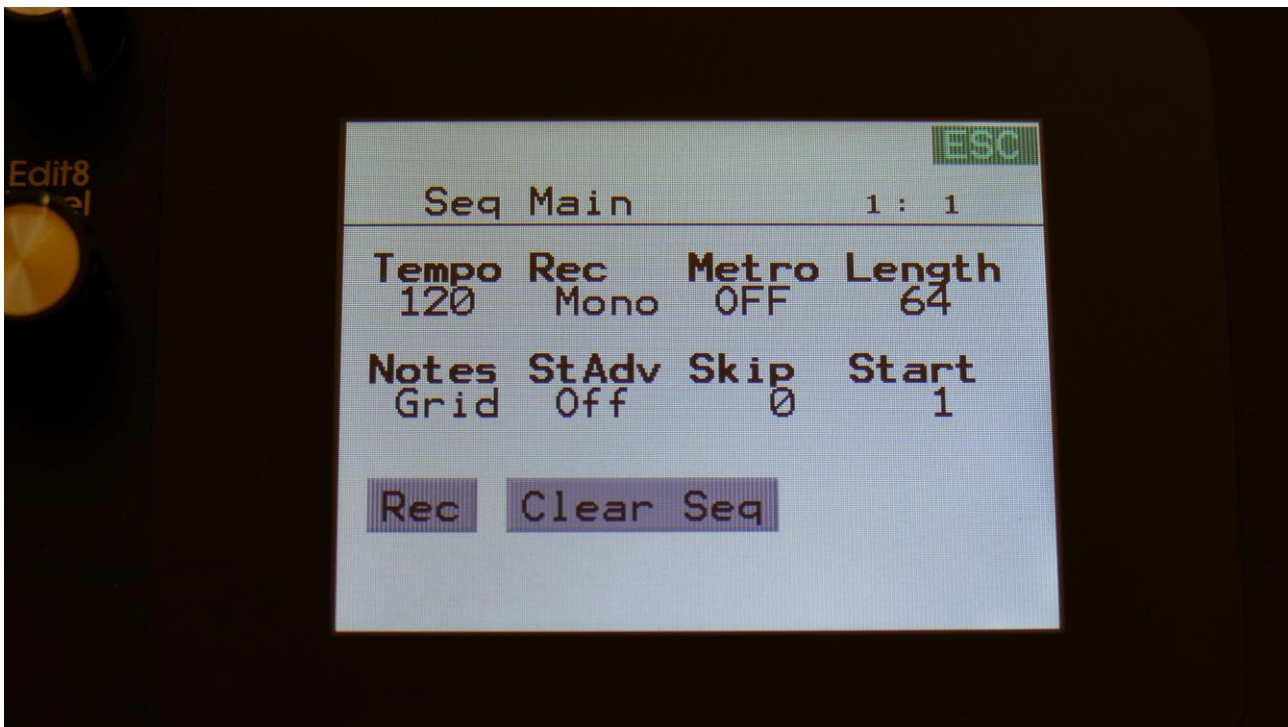
To exit to the Preset Select page, touch "ESC".

OR: Push and release the Func/Mute button, so that it lights up, and push step button 2/Seq Main, to access the Sequencer Main page.

Push step button 3/Note Trk to access the note track pages.

Push step button 4/Ctrl Trk to access the Controller Track pages. Push 3 times to access controller tracks 17-32.

Sequencer Main



The parameters of this page:

Tempo: Sets the sequencer tempo between 1 and 256 BPM. To make the sequencer sync to external MIDI gear, this must be set up in the “MOR>COMMON” section, described later in this manual.

Rec: Note Tracks Realtime Record Mode. Can be:

-**Mono:** The note tracks are recorded monophonically, with one step per position. Each step is positioned in a linear fashion, with step 1 on position 1, step 2 on position 2 and so on...

-**Poly:** The note tracks are recorded polyphonically. The positions of the steps are programmed every time a note is received. If more notes are played at the same time, or at the same position in time, more steps will have the same position. Only steps that are off can be recorded. Each step gets positioned at the moment, that it is recorded. Step 1 is recorded first, and positioned to the position in time of the recorded note. Then step 2 is recorded and positioned. And so on...

-**Audio:** When sequencer recording is started, the sample recorder is also started. Audio is recorded from the selected tracks start point, to the selected tracks end point. When recording stops, a sample trigger is placed at step 1, to trigger the sampling, and the audio track will immediately start to play back. See the “Audio Tracks” section later in this manual.

-AudCH: Chopped audio track. When the audio track is recorded, chop points are inserted at each sequencer step, according to the resolution setting of the selected track. See the “Audio Tracks” section later in this manual.

Metro: Metronome Off, On, Rec.

-Off: The metronome is off, and will make no sound.

-On: The metronome is on, and when the sequencer is playing, it will put out a click to audio output 1, every time note track 1 passes a 4/4 beat. Every time note track 1 starts over, it will make a click that is slightly higher pitched than the other clicks.

-Rec: When the sequencer is playing, and realtime rec is activated, it will put out a click to audio output 1, every time note track 1 passes a 4/4 beat. Every time note track 1 starts over, it will make a click that is slightly higher pitched than the other clicks.

Length: Global tracks length. Sets the step length for all the 4 Note tracks and 8 Controller tracks equal with one parameter. From 1 to 64 steps. Track lengths can be set individually for each track, in the separate track menu’s.

Notes: Sets if the notes of the Note tracks should be shown in a note grid (setting: Grid), or as value bars (setting: Value). This can also be switched on the note track pages.

StAdv: Step Advance on/off. When step recording a sequence, it will automatically advance to the next step of the selected track, if this is set to on.

Skip: Turning this up, will cause the sequencer to randomly skip some beats here and there. The more it is turned up, the more it will skip.
This affects all sequencer tracks at the same time.

Start: Global tracks start step. Sets the start step for all the 4 Note tracks and the 8 Controller tracks equal with one parameter. From step 1 (default) to step 64. Track steps can be set individually for each track, in the separate track menu’s.

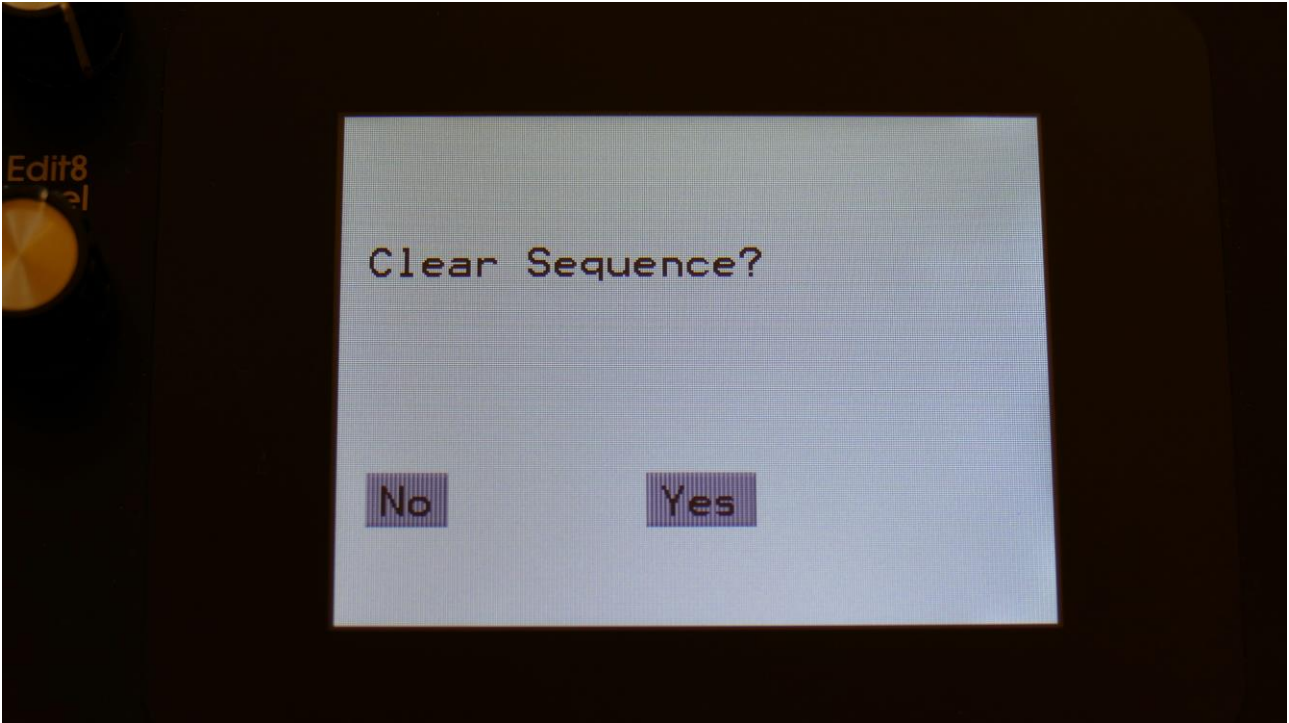
The buttons of this page:

Rec: This will toggle **realtime recording** on and off. **Realtime recording** can be switched on and off at any time, regardless of if the sequencer is playing or not. The sequencer must though play, to make it possible to input notes in realtime.

See “Realtime recording of notes” later in this section.

Clear Seq:

Touching this button, will make LD3 jump to this page:



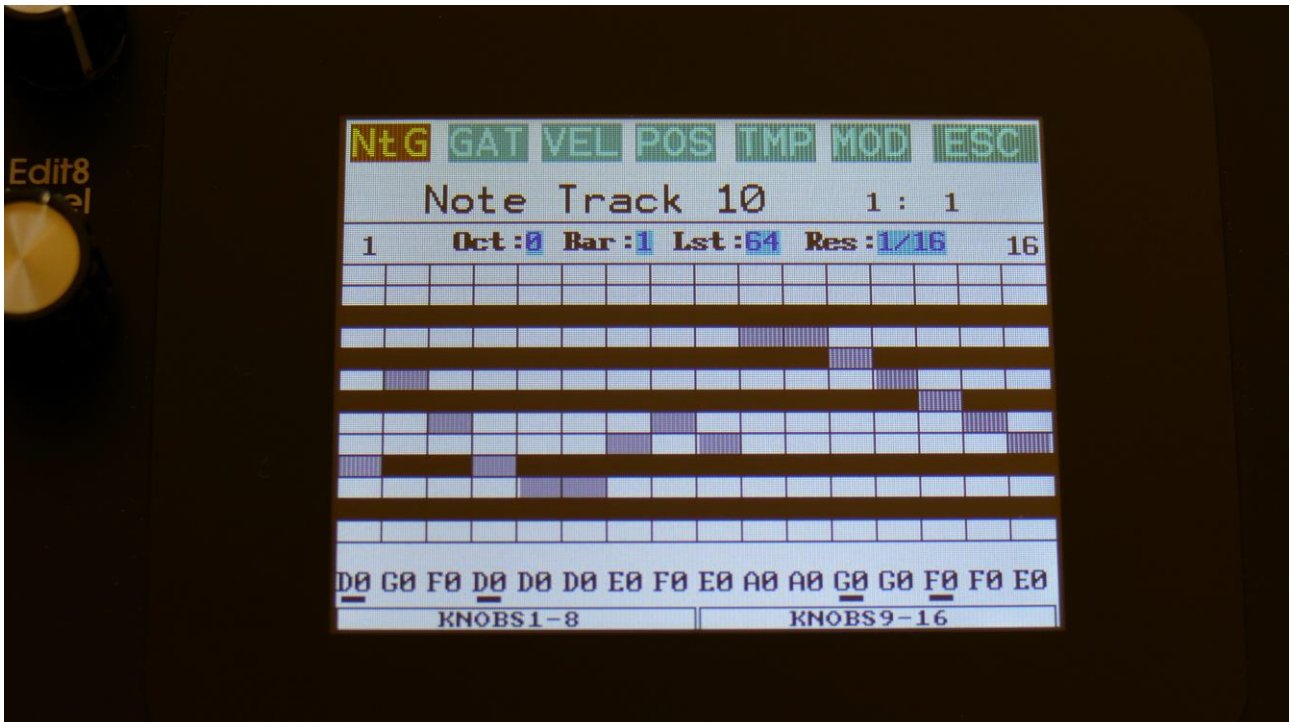
If you touch the “Yes” button here, you will clear all tracks of the current preset, and reset all values.

If you do not wish to do that, touch “No”.

Note Tracks

Note Subtrack

From the sequencer access page, touch NOTE TRACK 1-16, to enter the note steps edit page. OR: Push and release the Func/Mute button, so that it lights up, and push step button 3/Note Trk 1 time.



This will be the first page, you will enter, if the Notes parameter on the Seq Main page is set to Grid. At any time, touch NtG/NtV on the top navigation bar, to enter the note values page.

At most of this screen, you will see the note grid. Right above this, to the left and to the right, the first and last step numbers, of the steps that is currently shown in the note grid, are indicated.

Between these numbers, a few settable parameters is shown.

In the top left side, a small VU-meter is shown, that will indicate every time a note is played back.

Here, when neither KNOBS1-8 or KNOBS9-16 is black:

Edit Knob 1: Selects the octave to be viewed. The octave number is shown just above the note grid (marked **Oct**), and the lowest and highest note number is shown below and above the mini keyboard at the left.

Edit Knob 2: Selects the bar to be viewed. 4 bars can be viewed, each showing 16 steps, for a total of 64 steps. The bar number is shown just above the note grid (marked **Bar**), and the first and the last step number viewed, is shown just above the note grid, to the right and the left.

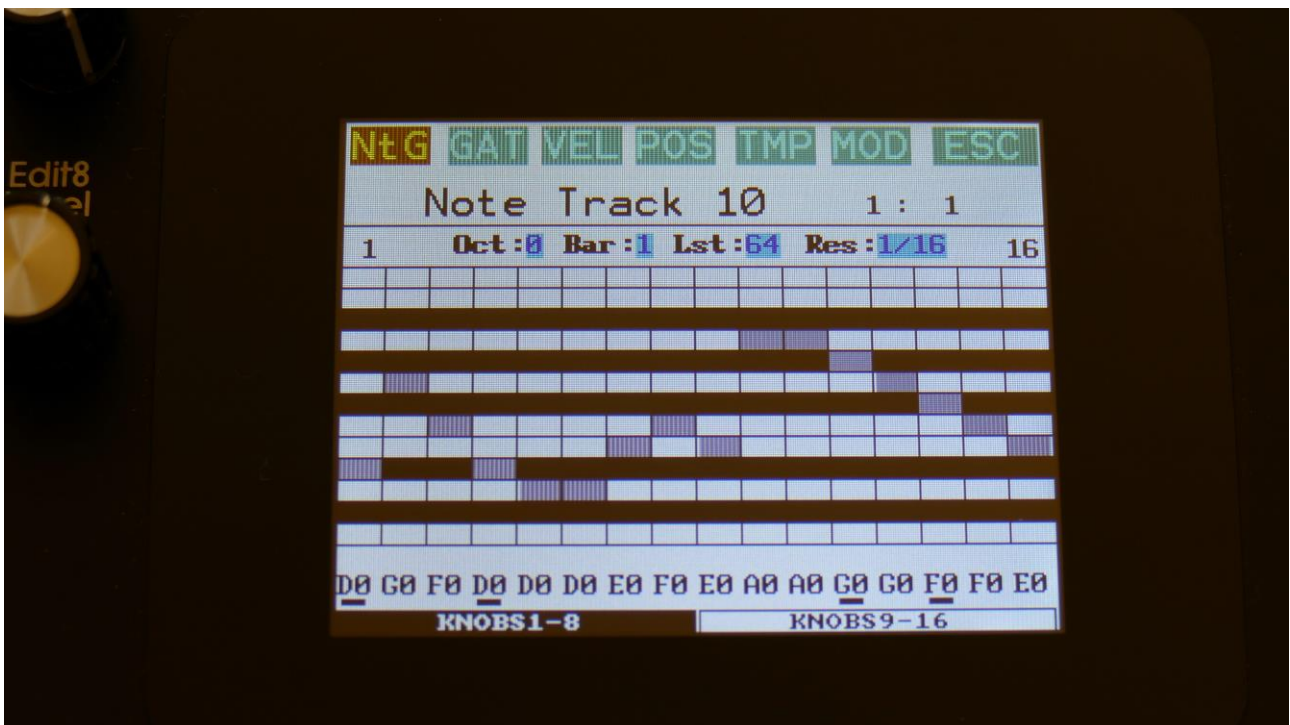
Edit Knob 3: Sets the last step that will be played back on this track, before it loops back, and plays the track again. Range: 1 to 64 steps. Value is shown as **Lst**.

The Last step can be set globally for all tracks at the Seq Main page.

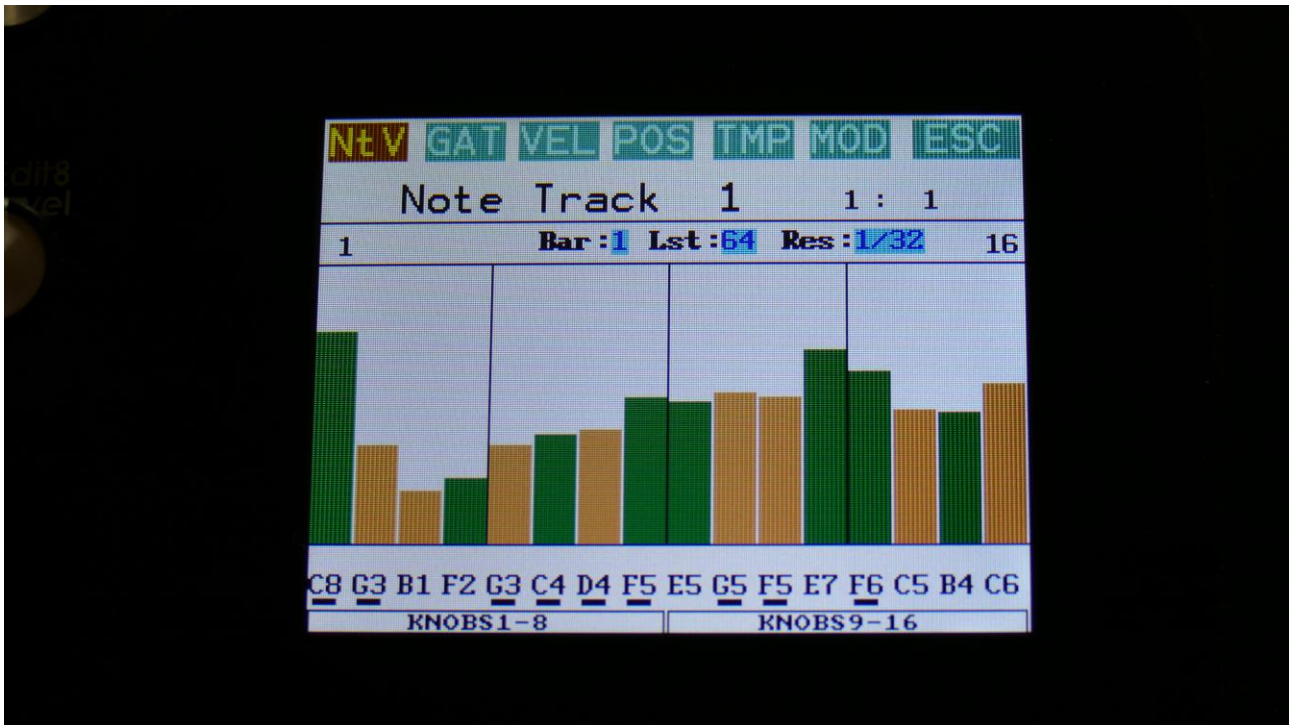
Edit Knob 4: Sets the track resolution. Values: 1/64, 1/48, 1/32, 1/24, 1/16, 1/12, 1/8, 1/6, 1/4, 1/2. Value is shown as **Res**.

The note values for each step is shown right below the note grid. If a black underscore is shown below the number, it means that it is a # note. Note values for each step can be changed, simply by touching the note grid.

Values can also be changed, using edit knob 1 to 8. In the bottom of the page, touch the field named “Knobs 1-8”, to edit the first 8 steps, using the edit knobs, or touch the field named “Knobs 9-16”, to edit the last 8 steps, using edit knob 1 to 8.



The note grid view are a bit limited, since you can only view one octave at a time. It is also possible to view the whole octave range at one time, by switching to the value bar view mode. Simply touch the NtG square in the upper left corner of the screen, or go to the sequencer main page, and change the Notes parameter to Value. The note values page will now look like this:

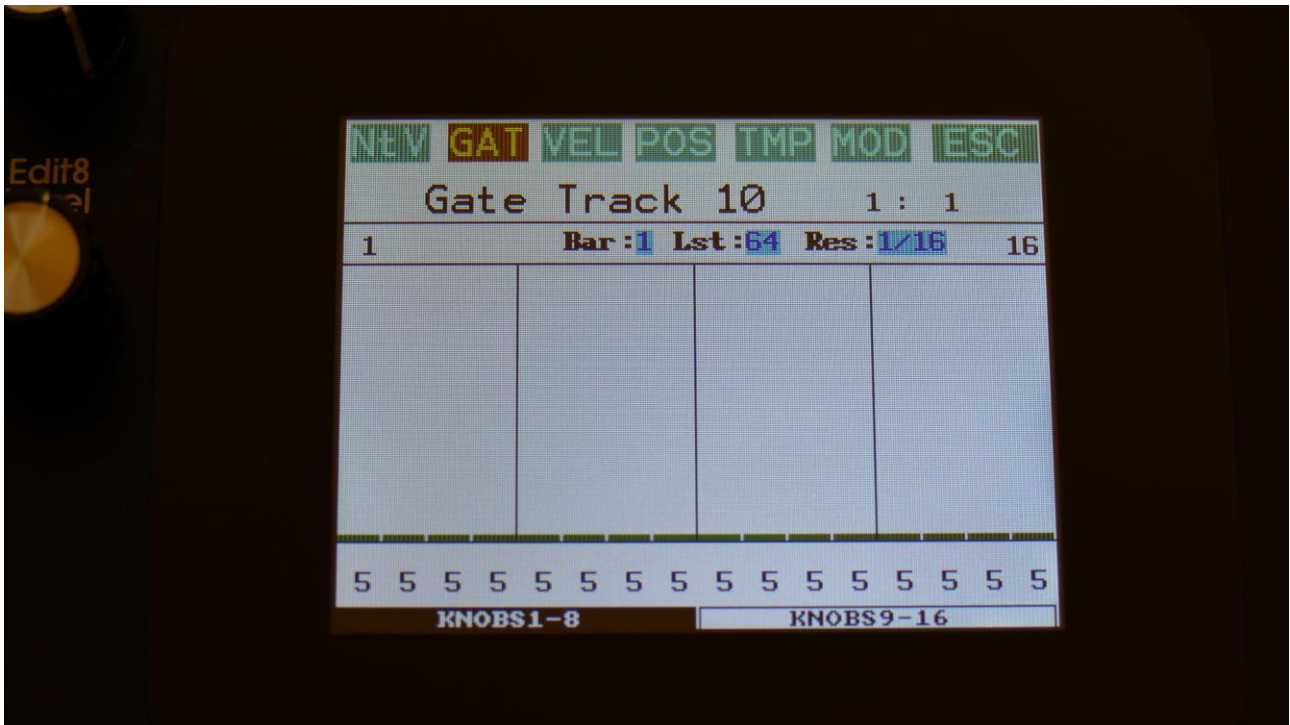


The knob functions are the same on this page, as on the note grid page. Note values can still be entered by touching the value bars, or by the edit knobs.

Steps that are on has a green value bar, steps that are off, has a yellow/red value bar.

Gate Time Subtrack

Touch GAT in the top navigation bar, to enter the gate time steps page. OR: Push and release the Func/Mute button, so that it lights up, and push step button 3/Note Trk 2 times.



The gate time values for each step is shown as bars in the value grid, and as number right below the value grid. If a black underscore is shown below the number, it means that it is above 100.

Gate time values for each step can be changed, simply by touching the value grid.

Values can also be changed, using edit knob 1 to 8. In the bottom of the page, touch the field named "Knobs 1-8", to edit the first 8 steps, using the edit knobs, or touch the field named "Knobs 9-16", to edit the last 8 steps, using edit knob 1 to 8.

When neither KNOBS1-8 or KNOBS9-16 are black:

Edit Knob 1: Has no function.

Edit Knob 2: Selects the bar to be viewed. 4 bars can be viewed, showing each 16 steps, for a total of 64 steps. The bar number is shown just above the note grid (marked **Bar**), and the first and the last step number viewed, is shown just above the note grid, to the right and the left.

Edit Knob 3: Sets the last step that will be played back on this track, before it loops back, and plays the track again. Range: 1 to 64 steps. Value is shown as **Lst**.

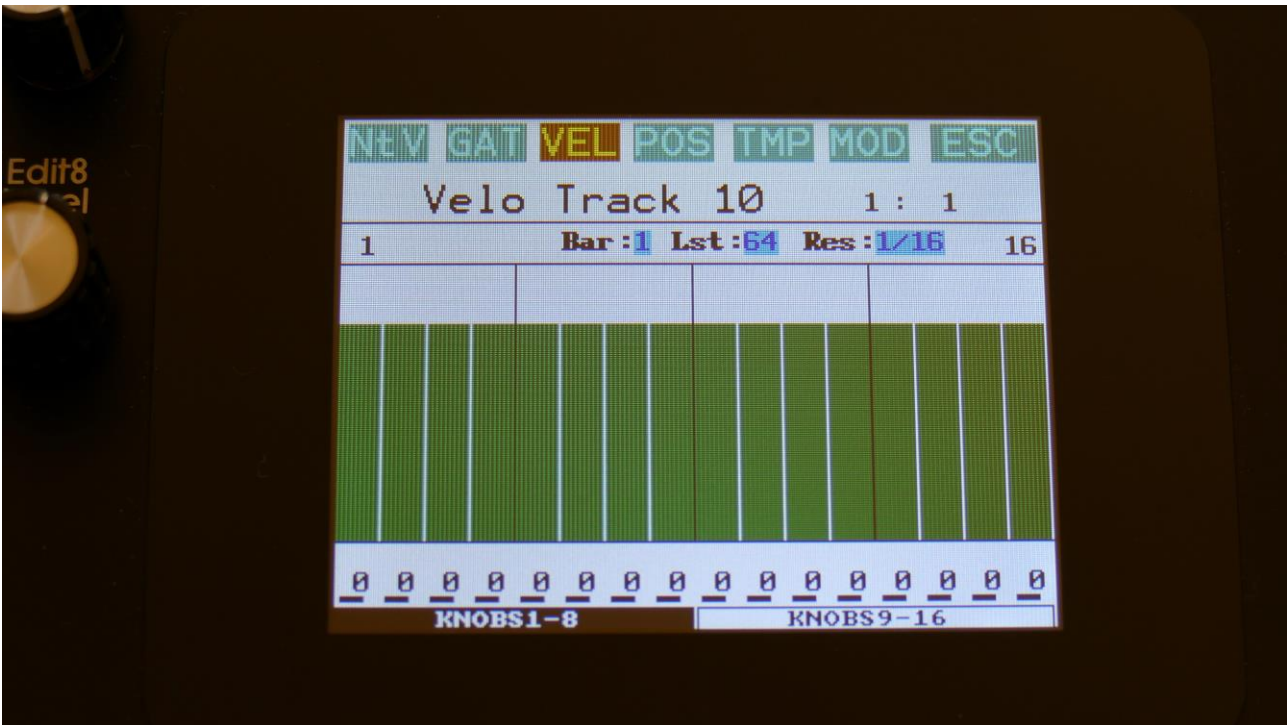
The Last step can be set globally for all tracks at the Seq Main page.

Edit Knob 4: Sets the track resolution. Values: $1/64$, $1/48$, $1/32$, $1/24$, $1/16$, $1/12$, $1/8$, $1/6$, $1/4$, $1/2$. Value is shown as **Res**.

Touching "ESC" will exit you to the Sequencer access page.

Velocity Subtrack

Touch VEL in the top navigation bar, to enter the velocity steps page. OR: Push and release the Func/Mute button, so that it lights up, and push step button 3/Note Trk 3 times.



The velocity values for each step is shown as bars in the value grid, and as number right below the value grid. If a black underscore is shown below the number, it means that it is above 100. Velocity values for each step can be changed, simply by touching the value grid.

Values can also be changed, using edit knob 1 to 8. In the bottom of the page, touch the field named "Knobs 1-8", to edit the first 8 steps, using the edit knobs, or touch the field named "Knobs 9-16", to edit the last 8 steps, using edit knob 1 to 8.

When neither KNOBS1-8 or KNOBS9-16 are black:

Edit Knob 1: Has no function.

Edit Knob 2: Selects the bar to be viewed. 4 bars can be viewed, showing each 16 steps, for a total of 64 steps. The bar number is shown just above the values field (marked **Bar**), and the first and the last step number viewed, is shown just above the values field, to the right and the left.

Edit Knob 3: Sets the last step that will be played back on this track, before it loops back, and plays the track again. Range: 1 to 64 steps. Value is shown as **Lst**.

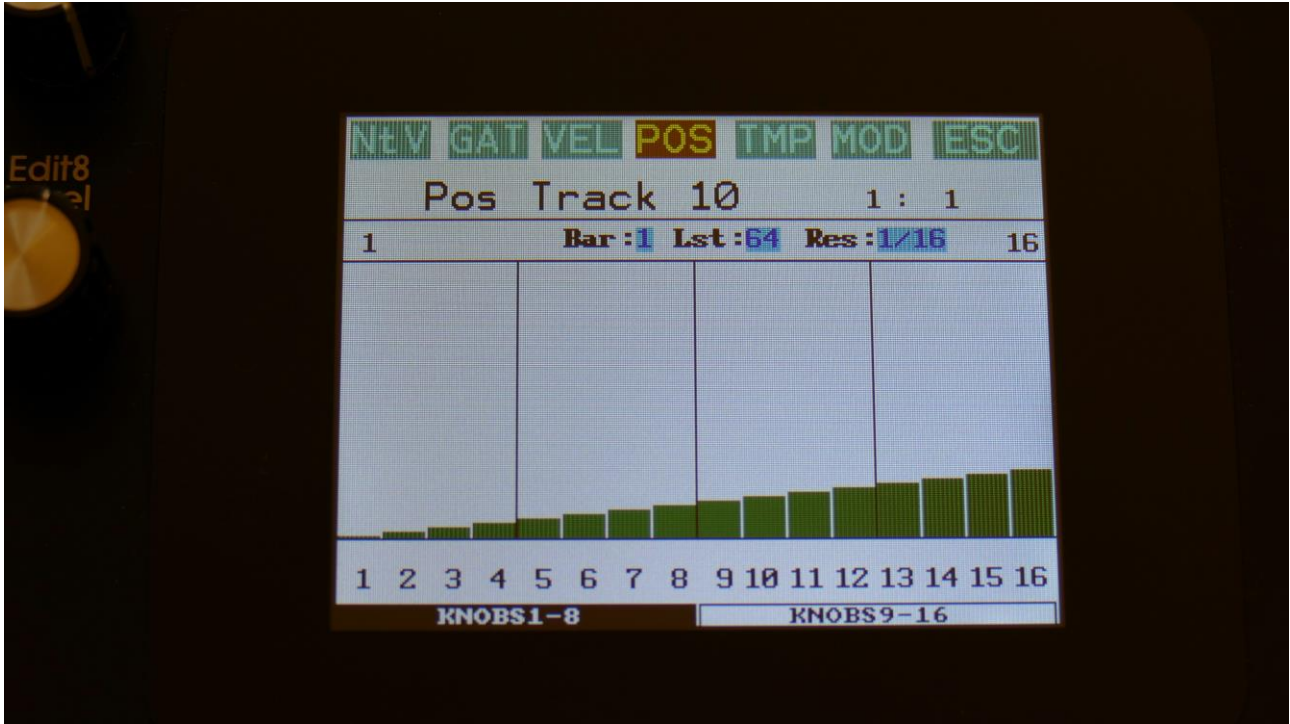
The Last step can be set globally for all tracks at the Seq Main page.

Edit Knob 4: Sets the track resolution. Values: 1/64, 1/48, 1/32, 1/24, 1/16, 1/12, 1/8, 1/6, ¼, ½. Value is shown as **Res**.

Touching “ESC” will exit you to the Sequencer Modules page.

Position Subtrack

Touch POS in the top navigation bar, to enter the position steps page:



Unlike many other step sequencers, the position for each step on the LD3 note tracks, can be set. For traditional step sequencing, you would leave it at its default setting, when step 1 is played back at position 1, step 2 is played back at position 2 and so on.

But the note sequencer in LD3 can go further than that. Step 2 can be set to play back at position 31, if desired, just to deform a track, or experiment with it, or more steps can be set to play back at the same position, for polyphonic sequencing.

Templates (explained in the next section) are available, to force the position to make the track play forwards, backwards, random, alternating and in many other patterns, if you do not like to set this up yourself.

The position for each step is shown as bars in the value grid, and as number right below the value grid. Positions for each step can be changed, simply by touching the value grid.

Values can also be changed, using edit knob 1 to 8. In the bottom of the page, touch the field named "Knobs 1-8", to edit the first 8 steps, using the edit knobs, or touch the field named "Knobs 9-16", to edit the last 8 steps, using edit knob 1 to 8.

When neither KNOBS1-8 or KNOBS9-16 are black:

Edit Knob 1: Has no function.

Edit Knob 2: Selects the bar to be viewed. 4 bars can be viewed, showing each 16 steps, for a total of 64 steps. The bar number is shown just above the values field (marked **Bar**), and the first and the last step number viewed, is shown just above the values field, to the right and the left.

Edit Knob 3: Sets the last step that will be played back on this track, before it loops back, and plays the track again. Range: 1 to 64 steps. Value is shown as **Lst**.

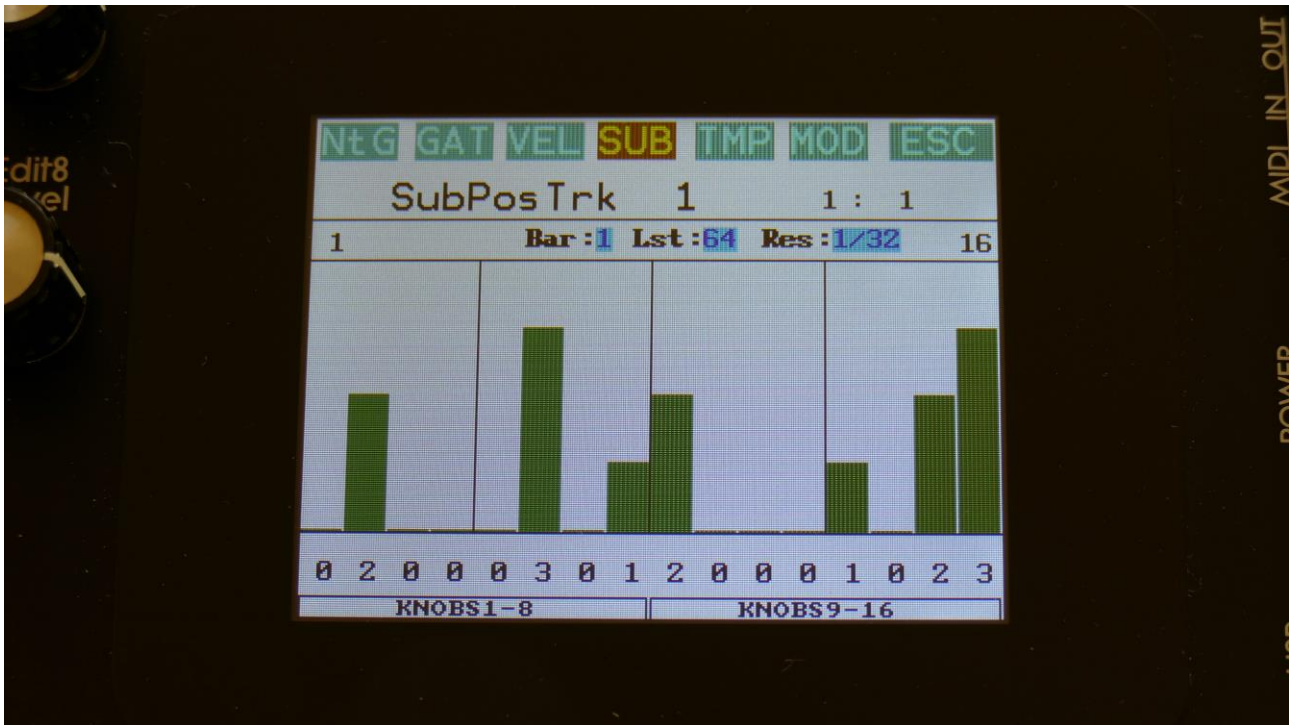
The Last step can be set globally for all tracks at the Seq Main page.

Edit Knob 4: Sets the track resolution. Values: $1/64$, $1/48$, $1/32$, $1/24$, $1/16$, $1/12$, $1/8$, $1/6$, $1/4$, $1/2$. Value is shown as **Res**.

Touching "ESC" will exit you to the Sequencer Modules page.

Sub Position Subtrack

When on the Position subtrack page, touch POS in the top navigation bar an extra time, to enter the sub position subtrack:



Here a value between 0 and 3 can be added to the step position for “micro-timing”.

The sub position for each step is shown as bars in the value grid, and as number right below the value grid. Sub positions for each step can be changed, simply by touching the value grid. Values can also be changed, using edit knob 1 to 8. In the bottom of the page, touch the field named “Knobs 1-8”, to edit the first 8 steps, using the edit knobs, or touch the field named “Knobs 9-16”, to edit the last 8 steps, using edit knob 1 to 8.

When neither KNOBS1-8 or KNOBS9-16 are black:

Edit Knob 1: Has no function.

Edit Knob 2: Selects the bar to be viewed. 4 bars can be viewed, showing each 16 steps, for a total of 64 steps. The bar number is shown just above the values field (marked **Bar**), and the first and the last step number viewed, is shown just above the values field, to the right and the left.

Edit Knob 3: Sets the last step that will be played back on this track, before it loops back, and plays the track again. Range: 1 to 64 steps. Value is shown as **Lst**.

The Last step can be set globally for all tracks at the Seq Main page.

Edit Knob 4: Sets the track resolution. Values: $1/64$, $1/48$, $1/32$, $1/24$, $1/16$, $1/12$, $1/8$, $1/6$, $1/4$, $1/2$. Value is shown as **Res**.

Touching "ESC" will exit you to the Sequencer Modules page.

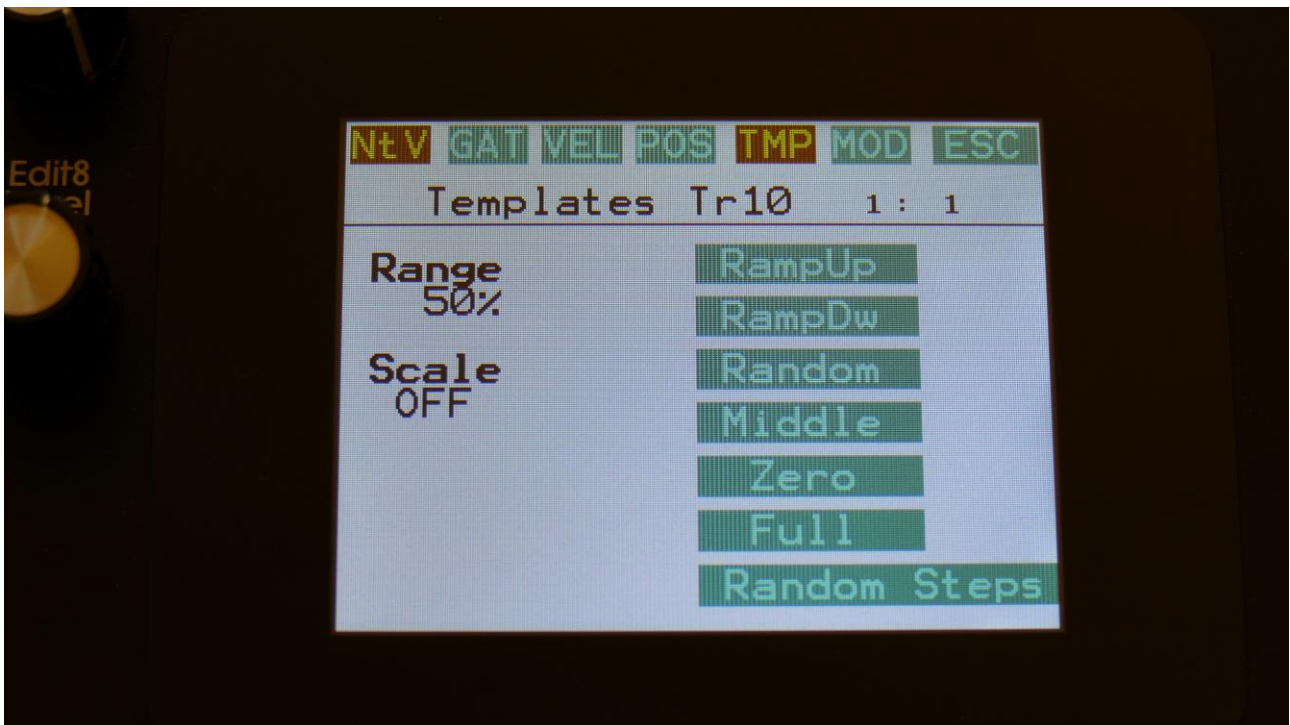
Templates and Note Scales

Templates are available to make it easier to create ramps on a track, and to randomize things. For the position track templates are available for forwards, backwards, alternating, randomized and polyphonic playback setup.

On this page note scales are also available, for users that thinks that harmonic tones works, and for swapping around keys for experimentation.

When on either the Note, Gate, Velo or position subtracks, touch TMP on the top navigation bar, to enter the templates/note scales page.

If you enter the templates pages from the note, gate or velocity page, it will look like this:



First turn Edit Knob 1, to set the range. This will set, how much the template routine will affect the values, that are already on the selected subtrack. When this is set to 100%, the template function will be absolute. Then touch any of the template boxes in the right side of the screen.

The templates work from step 1 to the last step that is set. Values after the last step are not affected.

Available templates:

- RampUp**: Ramp Up. Will lower values on steps, that are placed before last step divided by 2, and raise values on steps, placed after last step divided by 2.
- RampDw**: Ramp Down. Will raise values on steps, that are placed before last step divided by 2, and lower values on steps, placed after last step divided by 2.
- Random**: Will randomize all steps between step 1 and last step, to a degree set by the range parameter.
- Middle**: Will move any values closer to the middle value.
- Zero**: Will lower any values. –Transpose a note track down.
- Full**: Will raise any values. –Transpose a note track up.

Random Steps:

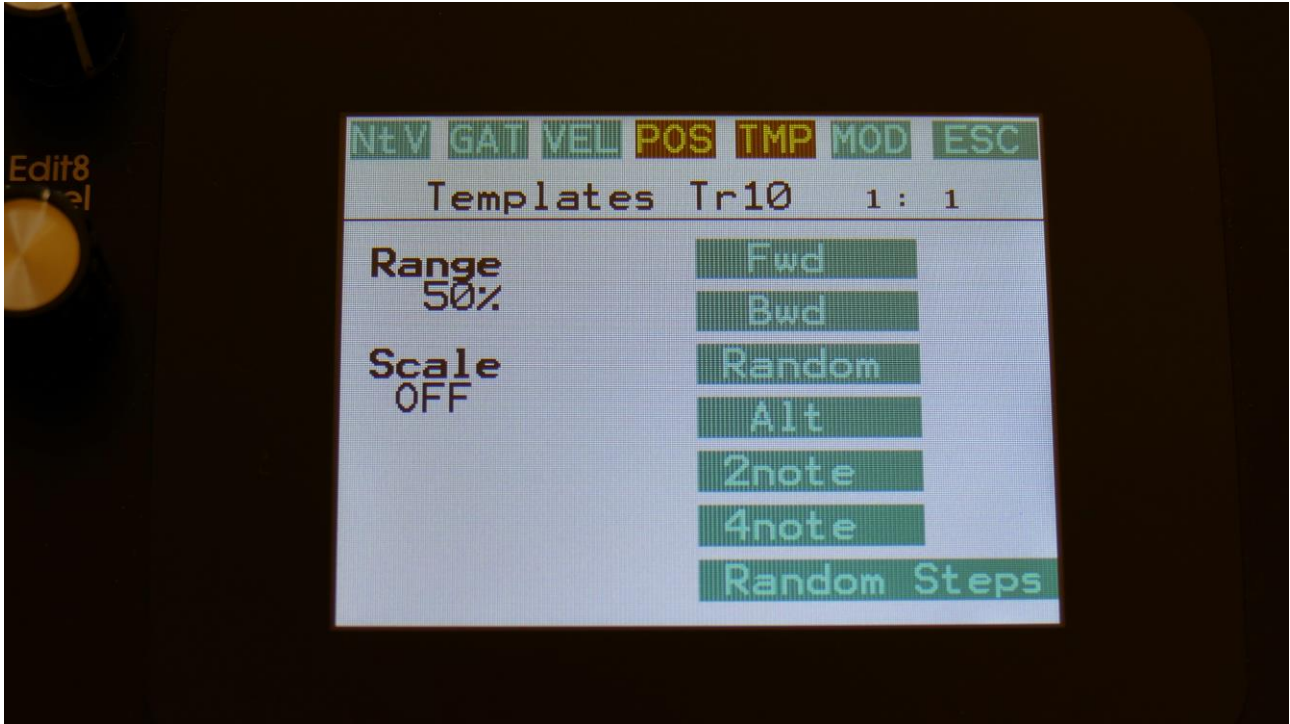
Will randomize the note steps on's/off's. The more that the Range parameter is turned up, the more notes it will set to on.

Scale: Adds a scale to the note track output in realtime, when adjusting Edit Knob 5. Available scales are:

- Off**: No scales are added to the note track.
- Cmaj**: Forces all notes outputted from the track, to the Cmajor scale.
- Dmaj**: Forces all notes outputted from the track, to the Dmajor scale.
- Emaj**: Forces all notes outputted from the track, to the Emajor scale.
- Fmaj**: Forces all notes outputted from the track, to the Fmajor scale.
- Gmaj**: Forces all notes outputted from the track, to the Gmajor scale.
- Amaj**: Forces all notes outputted from the track, to the Amajor scale.
- Bmin**: Forces all notes outputted from the track, to the Bmajor scale.
- Cmin**: Forces all notes outputted from the track, to the Cminor scale.
- Dmin**: Forces all notes outputted from the track, to the Dminor scale.
- Emin**: Forces all notes outputted from the track, to the Eminor scale.
- Fmin**: Forces all notes outputted from the track, to the Fminor scale.
- Gmin**: Forces all notes outputted from the track, to the Gminor scale.
- Amin**: Forces all notes outputted from the track, to the Aminor scale.
- Bmin**: Forces all notes outputted from the track, to the Bminor scale.
- Swap**: Will swap notes below F inside an octave, with notes above F, and vice versa.
- BIWh**: Will swap white keys with black and vice versa.

Templates for the Position Subtrack

When entering the templates page from the Position subtrack, it will look like this:



Turn Edit Knob 1, to set the range. This will set, how much the template routine will affect the values, that are already on the selected subtrack. When this is set to 100%, the template function will be absolute.

The templates work from step 1 to the last step that is set. Values after the last step are not affected.

Available templates:

- Fwd**: Forwards. The more the Range parameter is turned up, when touching this, the more the track will be forced to play back forwards.
- Bwd**: Backwards. The more the Range parameter is turned up, when touching this, the more the track will be forced to play back backwards.
- Random**: Will randomize all steps between step 1 and last step, to a degree set by the range parameter.
- Alt**: Alternate. The more the Range parameter is turned up, when touching this, the more the track will be forced to play back alternating.

-2note: 2 Note Poly. The more the Range parameter is turned up, when touching this, the more the track will be forced to play back 2 note polyphonic.

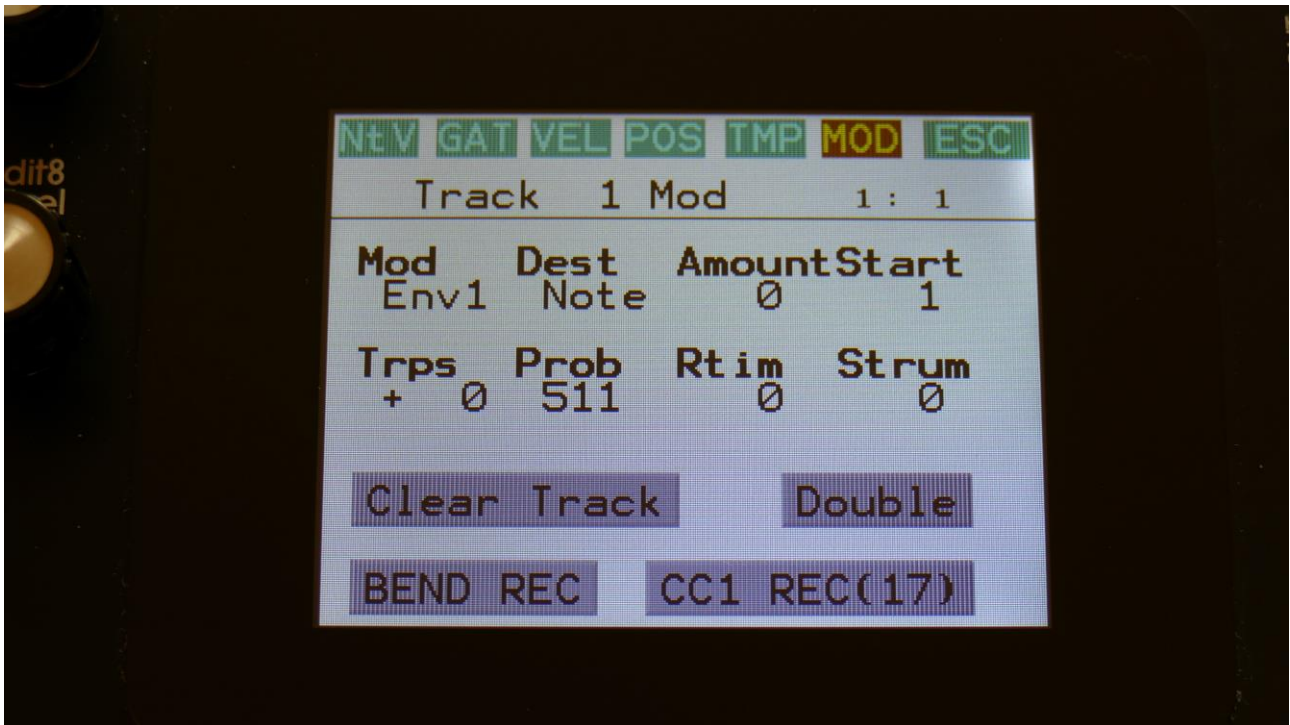
-4note: 4 Note Poly. The more the Range parameter is turned up, when touching this, the more the track will be forced to play back 4 note polyphonic.

Random Steps:

Will randomize the note steps on's/off's. The more that the Range parameter is turned up, the more notes it will set to on.

Note Track Mod Page

Touch MOD in the top navigation bar, to enter this page. OR: Push and release the Func/Mute button, so that it lights up, and push step button 3/Note Trk 4 times.



Here you can set up note track modulation, the track Note Randomizer, the start step, make the track strum, and transpose the track. It is also possible to clear or double the track.

The parameters of this page:

Mod: Note Track modulation source. Any LD3 modulation source can be selected to modulate the track. For a complete list of modulation sources, see the list in the start of the synth part section in this manual.

Dest: Track modulation destination. Selects what should be modulated on the track. Choices are:

-**Note:** The note numbers of the track will be modulated.

-**Gate:** The gate times of the track will be modulated.

-**Dly:** The delay times of the track will be modulated.

-**Swing:** Swing modulation will be applied to the track. This is a bit more fine than the delay time modulation.

-**Velo:** The velocity values of the track will be modulated.

-**Pos:** The position subtrack positions will be modulated.

-**Strum:** The Strum parameter will be modulated.

Amount: Adjusts the amount of modulation that will be applied to the track.

Start: Sets the track start step. This will be the step, the track starts at, when hitting the Start/Stop button, or when entering this preset from another preset, while the sequencer is running. When selecting another preset, while the sequencer is running, the preset change will happen, when Note track 1 are at the step before the start step. Step 1 to 64 can be selected.

The Start step can be set globally for all tracks at the Seq Main page.

Trps: Track transpose. Will transpose the note numbers of the whole track up to 64 semitones up or down.

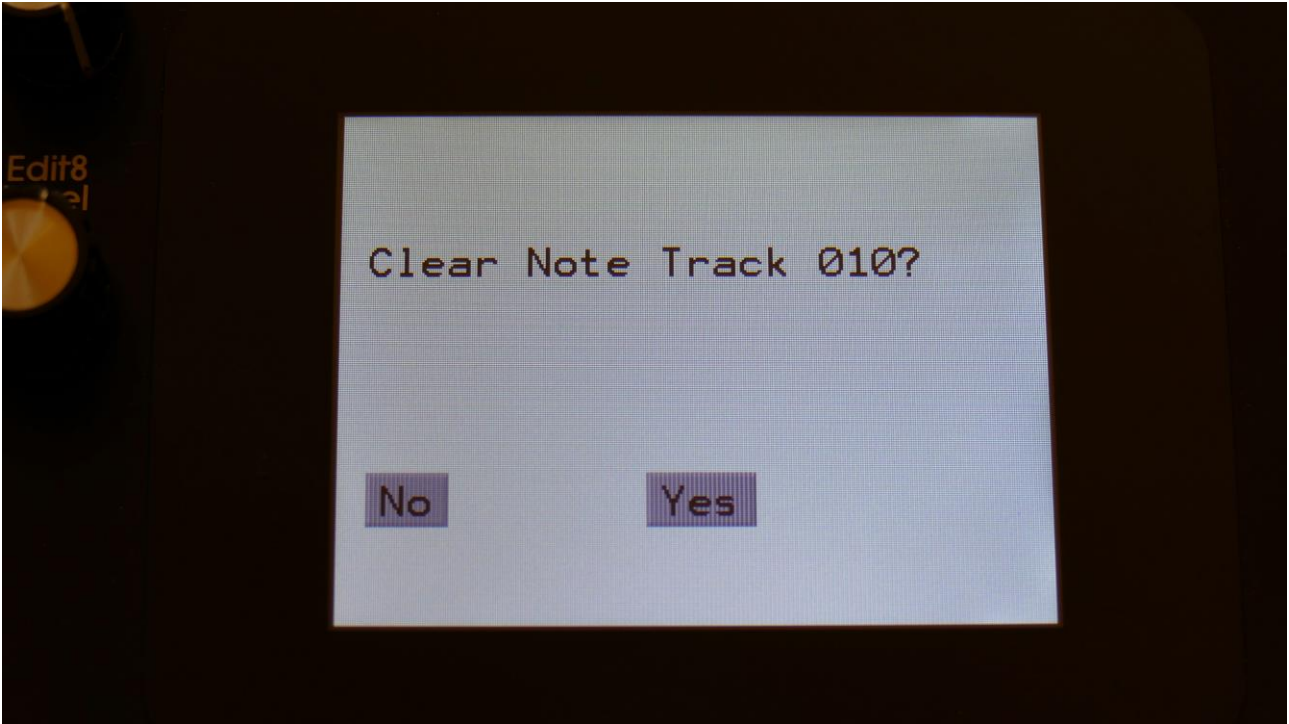
Prob: Track Note Randomizer Probability. When this is turned fully up, all programmed note steps of the track will play back. The more it is turned down, the less notes will play back. The notes the will not play back are selected randomly.

Rtim: Track Note Randomizer Time. When turned up, it will create random delays on the programmed note steps triggering.

Strum: When the track is playing more than one note at the same time, the more the Strum parameter is turned up, the longer the delay between each note will be.

Clear Note Track

From the Mod page, touch the “**Clear Track**” button, to enter this page:



Touching “Yes” from here, will initialize all step values and parameters of the currently selected note track, except for the Last Step parameter.

If you would not wish to do this, touch “No”, to exit to the track Mod page.

Double Note Track

Touch the “Double” touch button, to double the length of the track. All events from the original track, is copied to the new half. Useful if you need some extra variation in a track.

PLEASE NOTE: The track lengths cannot exceed 64 steps for the note tracks.

Realtime recording of notes

You can, at any time, hit the Steps/Part button, and switch note steps on or off, but sometimes you might like to just record something, that you play on a connected MIDI device, like a keyboard or a drum pad, on the touch screen keyboard, or on LD3's trigger buttons. For this, realtime recording is usable.

You can start realtime recording by holding down the Func/Mute button, while pressing the Start/Stop button. If the sequencer is already running, it will immediately enter realtime recording mode, and any note you play will be recorded. If the sequencer is stopped, you will have to push the Start/Stop button, to start it. Before it starts realtime recording, it will count in for 2 bars with audible metronome.

When realtime recording, the Start/Stop button will flash in time with the beat.

To switch off realtime recording, simply hit the same button combination again, so the Start/Stop button stops flashing.

LD3 has two different note realtime recording modes: Mono and poly.

In mono recording mode, it records the notes in a linear time fashion, like a traditional step sequencer: One note per step. The notes you play are placed at the position where the sequencer was playing back, at the moment that you played the note. If you play more than one note on the same position, only the last note you play are recorded.

If you play several parts at the same time, each part will be recorded on separate tracks, so more parts can be recorded to sound at the same time.

In poly recording mode, the position subtrack is modified every time you play a note to be recorded. The first note you play, will be placed at step one, and the position of step one, will be set to the position, that the sequencer was playing back, when you played the note. The second note you play, will be placed at step two, and the position of step two, will be set to the position, that the sequencer was playing back, when you played the note. And so on, until you have played 64 notes. Any step can be at any position, and therefore polyphonic recording and play back is possible.

The realtime recording mode is set on the Sequencer Main page, by adjusting the Rec parameter.

The choices are:

-Mono: Monophonic recording

-Poly: Polyphonic recording

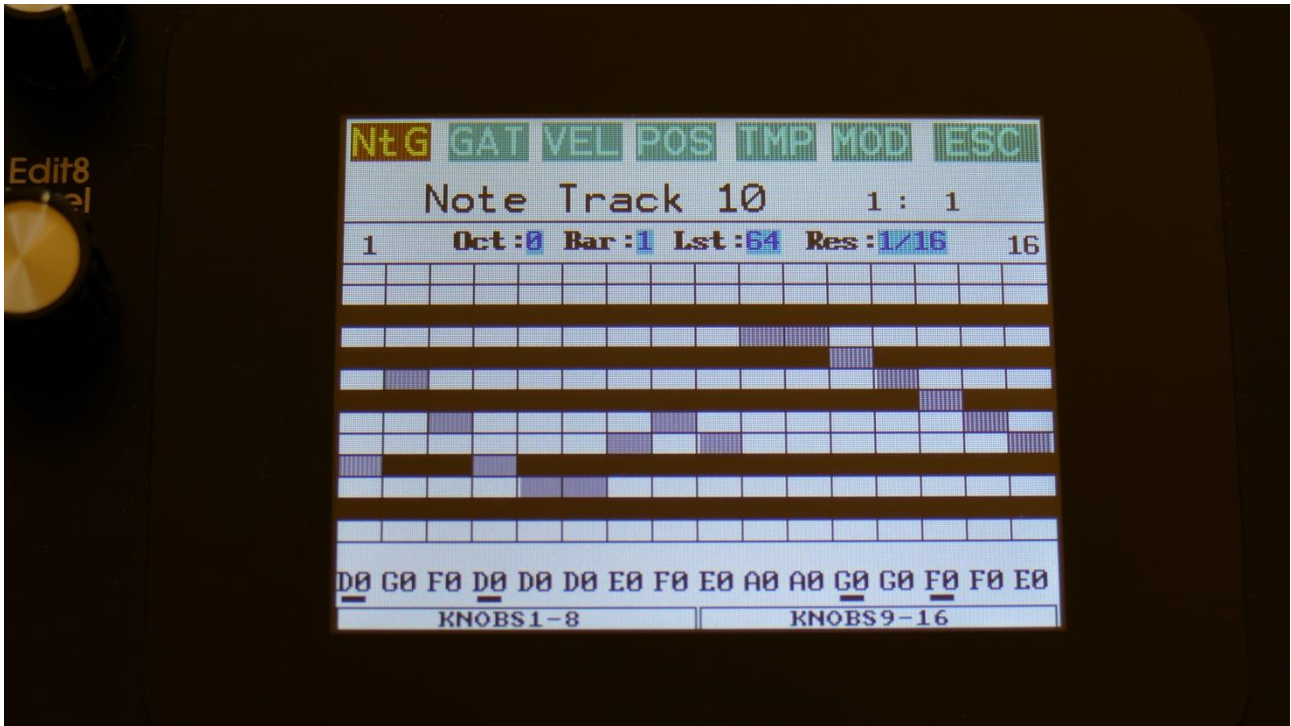
-Audio: Audio track recording (Explained a little bit later in this manual)

-AudCH: Chopped audio track recording (Explained a little bit later in this manual)

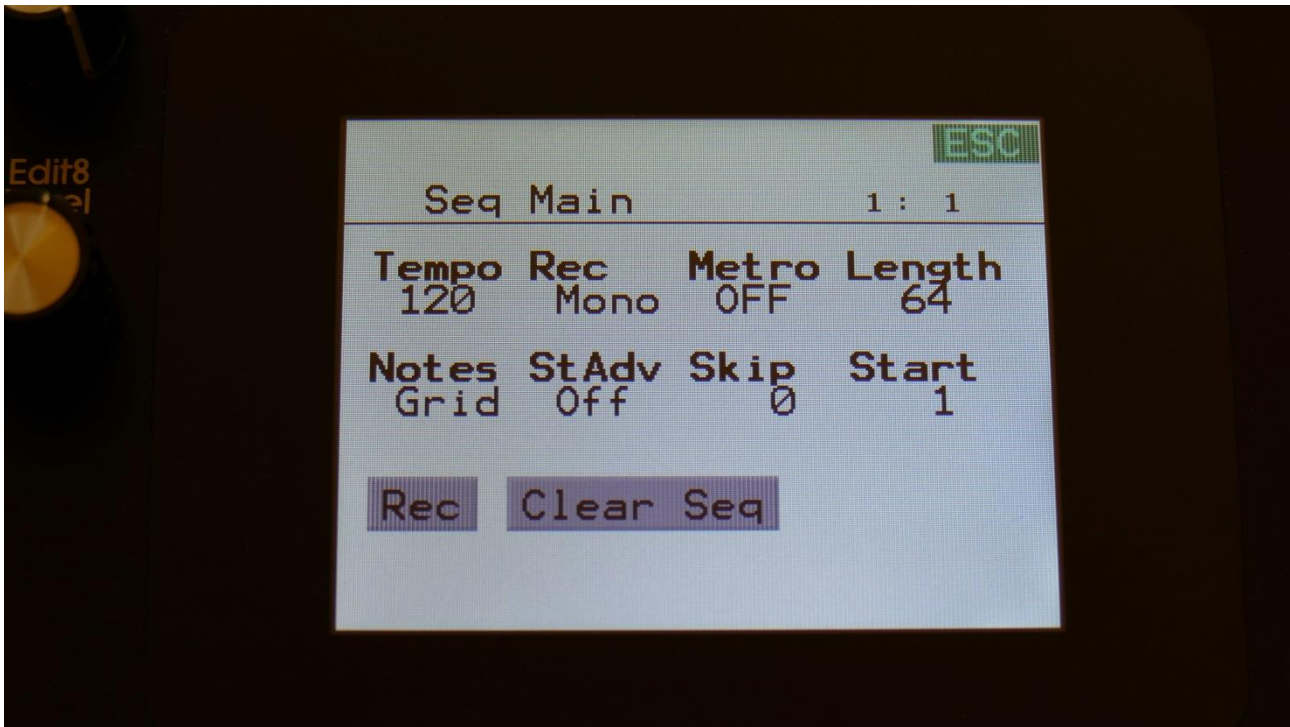


On the Sequencer Main page, it is also possible to switch realtime recording on and off, by touching the Rec touchbutton.

Step Recording of Notes



On top of the xox style and realtime recording methods, step recording of notes and controllers are also possible.



On the sequencer main page, it is by the StAdv (Step Advance) parameter possible to set, if the sequencer, when step recording, should automatically advance to the next step or not.

Rec mode can be set to either mono or poly.

To initiate step recording:

- Set LD3 in sequencer recording mode, by touching the REC field on the sequencer main page or on the preset select page, so that it turns red, or push and hold the Func/Mute button, while pressing the Start/Stop button.
- Start the sequencer playback, if desired. Step recording is possible no matter if the sequencer is running or not.
- Push and release the Steps/Part button, so that it lights up, and puts the 16 step buttons in step mode.
- Push the step button, that has the position for the step, that you would like to record. This will now flash.
- Play a note (or more in poly rec mode), on an attached MIDI device or on the touch screen keyboard. This will now be recorded. If any controller track is set in recording mode (explained in the next section), MIDI CC's and knob movements can also be recorded to specific steps this way.
- If Step Advance is set to on, LD3 will advance to the next step, every time it receives a note off.
- Else, now select another step to record.

If the sequencer is running, while step recording, the time you hold down a note, will be recorded.
If the sequencer is stopped, resolution will determine the note length.

Audio Track Recording

Each of the note tracks 1 to 8, can be set up to function as an audio track. One audio track can be recorded at a time.

Audio tracks can either be recorded as one whole piece (un-chopped), or they can be recorded as a number of smaller pieces, that can be re-arranged afterwards (chopped).

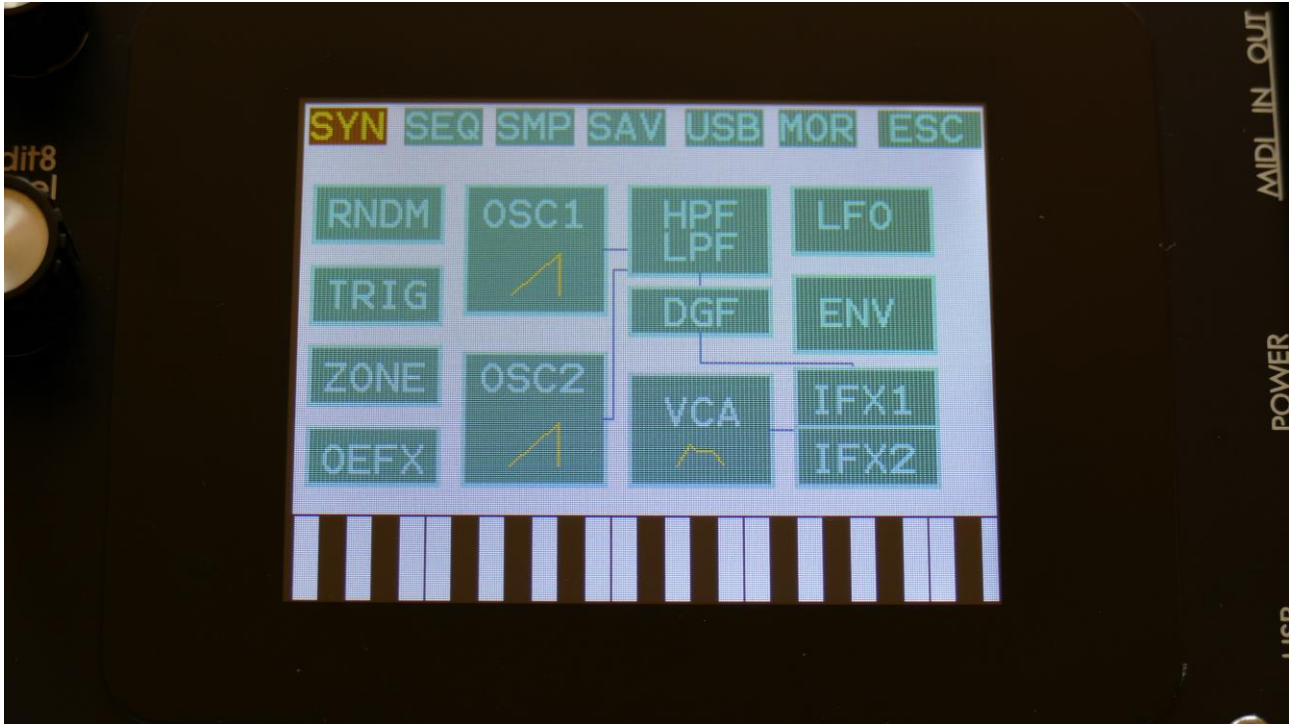
An audio track will be the same length, as the note track you are recording to, is set to. When recording chopped, the chops size will follow the resolution set on the note track. So if you plan to rearrange the track afterwards, it will probably be best to set it to as low a resolution as possible, while still getting the desired length.

When you just wants to record one whole piece, it is best to set the resolution as high as possible, so that you will get a longer recording – up to 32 bars, when resolution is set to $\frac{1}{2}$.

When recording a chopped audio track, the controller track that has the same number, as the note track you are recording, will be reserved by the system as a modulator for the oscillator chop select parameter. To rearrange the chopped pieces of the audio track, you will have to enter this controller track, and change the controller step values.

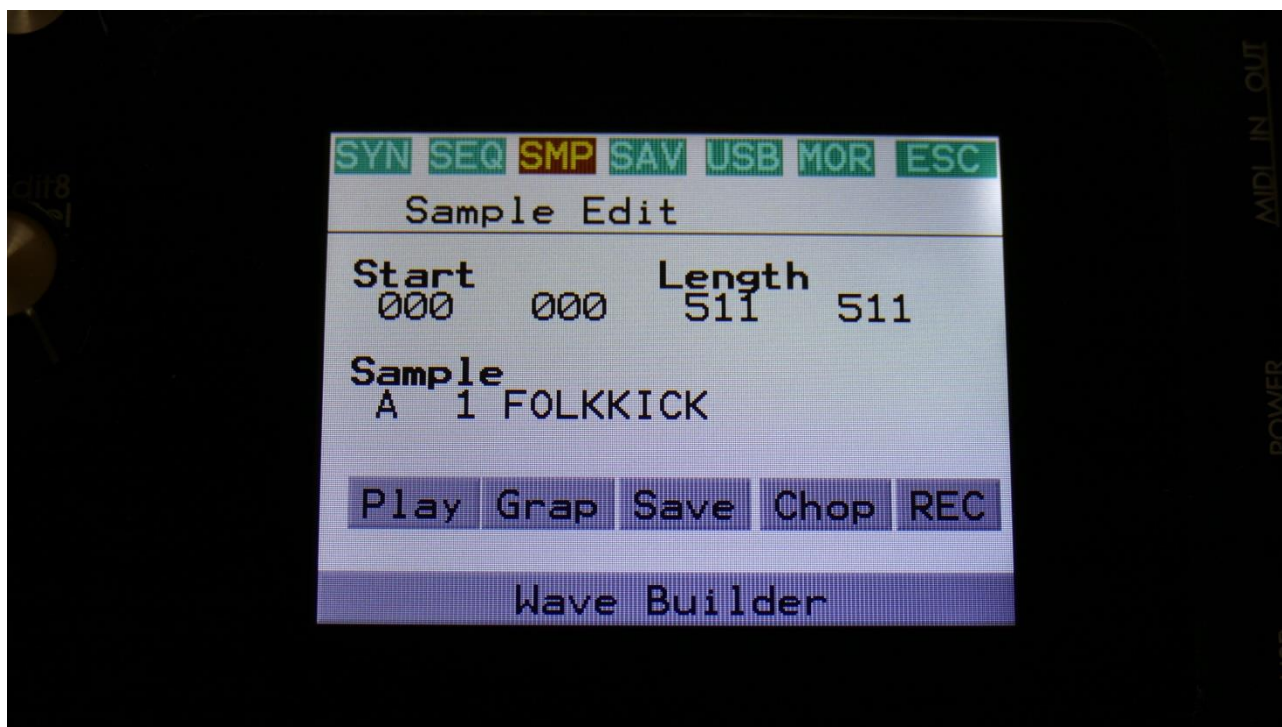
Audio track setup.

From the Preset select page, touch the EDIT field, to enter the synth main page.



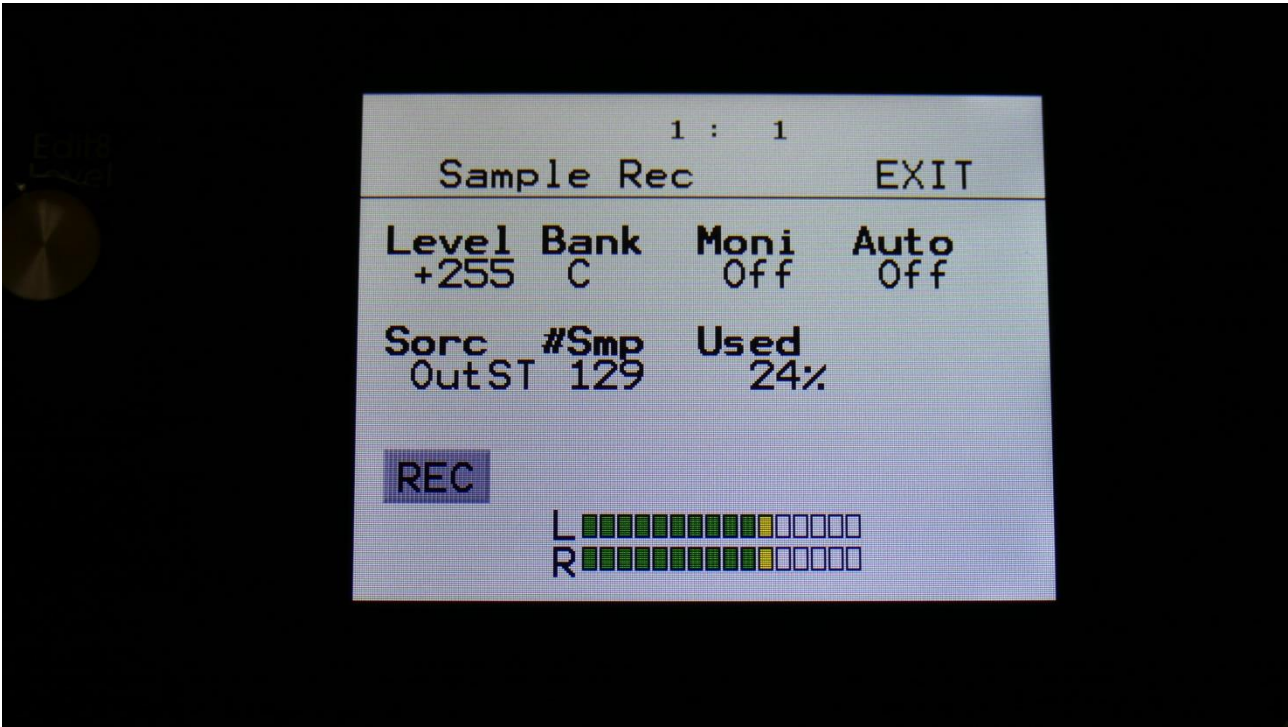
Touch SMP on the top navigation bar.

You should now enter this page:



Touch the REC touch button.

And you should get here:



OR, from any page:

-Push and release the Func/Mute button, so it is lighting up.

-Push step button 12, marked "Sample/Qrec".

This will also get you to this page, just much faster.

If you can already hear the source, that you would like to record, leave the Moni parameter Off. If you have the record source connected directly to any of LD3's audio inputs, and can't hear it now, switch the Moni parameter to On. If you only want to listen to the record source, while it is recording, and get the source signal switched off, at the moment recording stops, and the tracks starts to play back, set the Moni parameter to Rec.

Select the record source, by setting the Src parameter.

Choices are:

OutpL : Left audio output mono

OutpR : Right audio output mono

OutST : Left and right audio outputs stereo

In order to record the audio inputs, you must set up part 5 and 6, to let the audio from the analog filters of these part, go through to the outputs.

Now select the sample bank, that you would like to record to, setting the Bank parameter to bank A, B, C or D.

When choosing a bank, the **#Smp** parameter will show how many sampling that the selected bank already holds, and the **Used** parameter will show how many percent of the sample bank, that are already in use.

For optimal results, set the level parameter to +0, and adjust the level of the record source, so that the 2 VU-meters in the bottom of the screen does not get red.

You should only adjust the level parameter, to other values than +0, if it is not possible to adjust the level directly on the source.

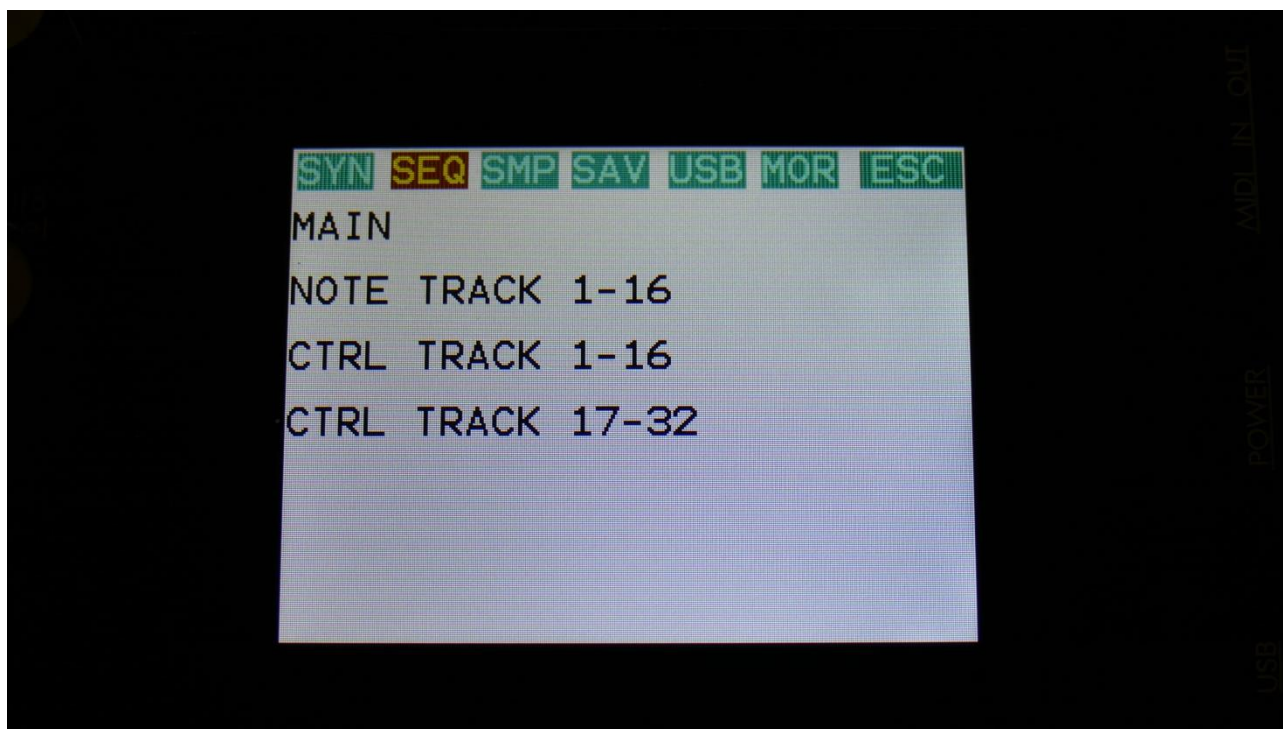
Now go to the sequencer note track page, either by:



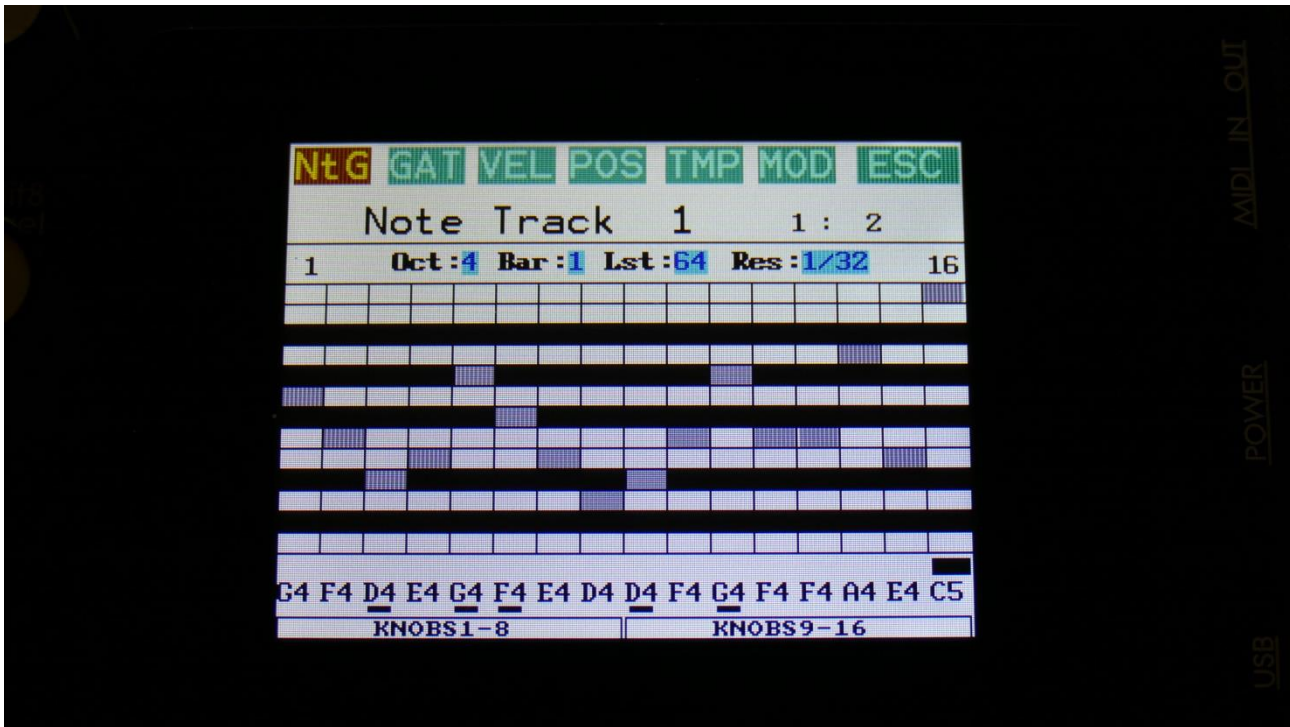
Exit the Sample Rec page, by touching EXIT in the upper right corner, to go back to the Sample Edit page.

Touch SEQ on the top navigation bar.

You should now have entered this page:



Touch NOTE TRACK 1-16, to enter the note track page.

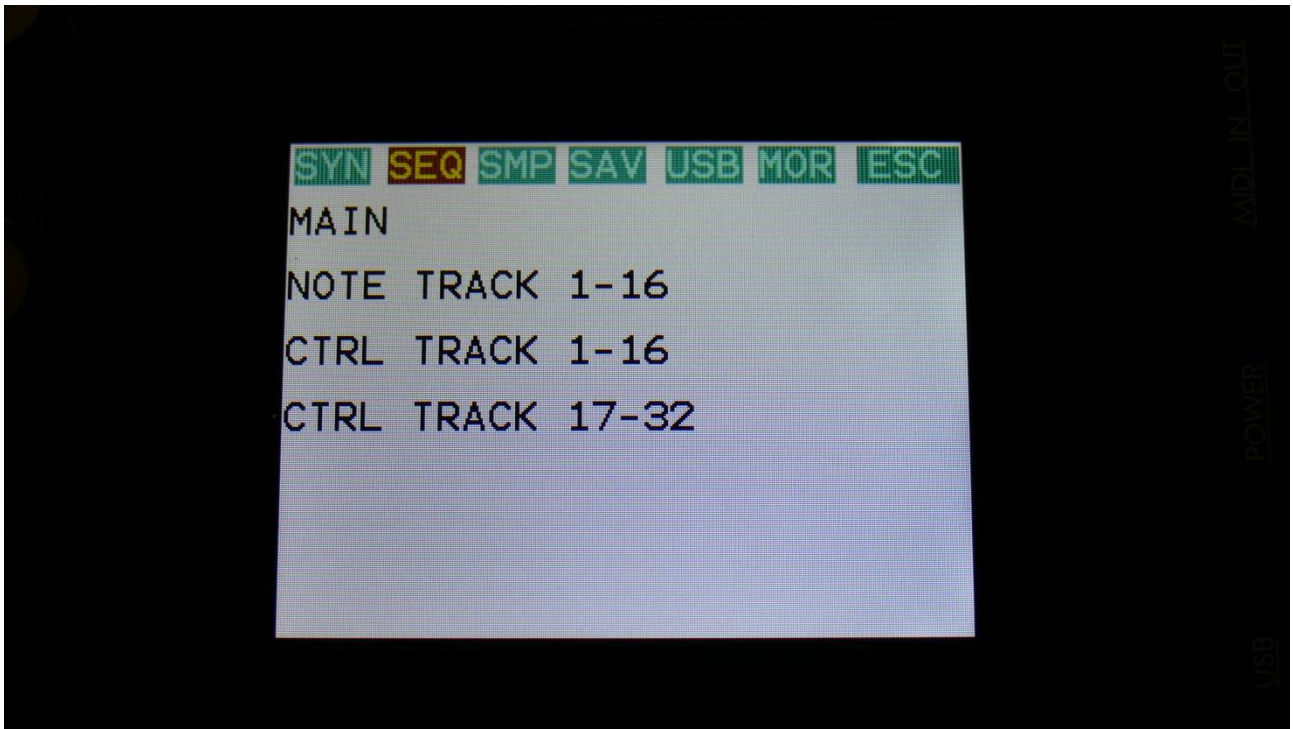


And the faster way to go from the Sample Rec page to the Note Track page:

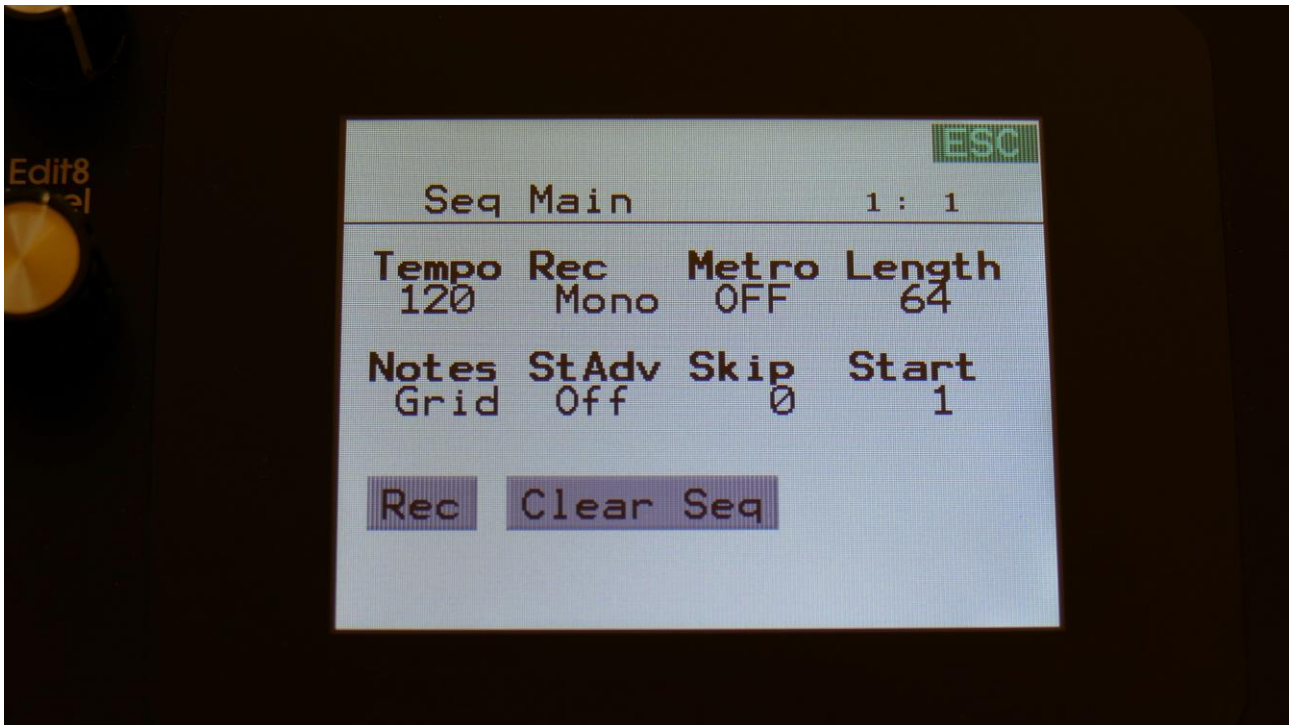
- Push and release the Func/Mute button, so it is lighting up.
- Push step button 3, marked "Note Trk".

- Now, select the note track, that you would like to record.
- Set the desired length and resolution.

Go to the sequencer main page, by touching ESC on the top navigation bar, to go back to the Seq overview page.



Touch MAIN.



And the faster way to go from the Note track page to the Sequencer Main page:

- Push and release the Func/Mute button, so it is lighting up.
- Push step button 2, marked "Seq Main".

Set the desired tempo.

Switch the metronome on, if desired.

Set the Rec parameter to either Audio, if you would like to record one long piece, or to AudCH, if you would like your recording chopped up.

To start recording, either touch the Rec touch button, or hold down the Func/Mute button, while pushing the Start/Stop button.

If the sequencer is already running, it will start recording the audio track, the next time the track starts over from step 1.

If the sequencer is not already running, push the Start/Stop button, to start it. LD3 will now count in for 2 bars, with the metronome audible, and then start recording.

As soon as it reaches the track last step, recording will stop, and the track will start to play back.

The Controller Tracks

LD3 has 32 controller tracks. Each of these only have one subtrack: The controller track values. Each track can be on up to 128 steps long, and a Smooth function can be turned on, that will automatically create values between the steps, from one step value to another, for smooth transitions between them.

Each controller track both creates an internal modulation source and a MIDI CC or Pitch Bend source. The MIDI CC's and Pitch Bend sources can be sent both to external MIDI gear, and internally.

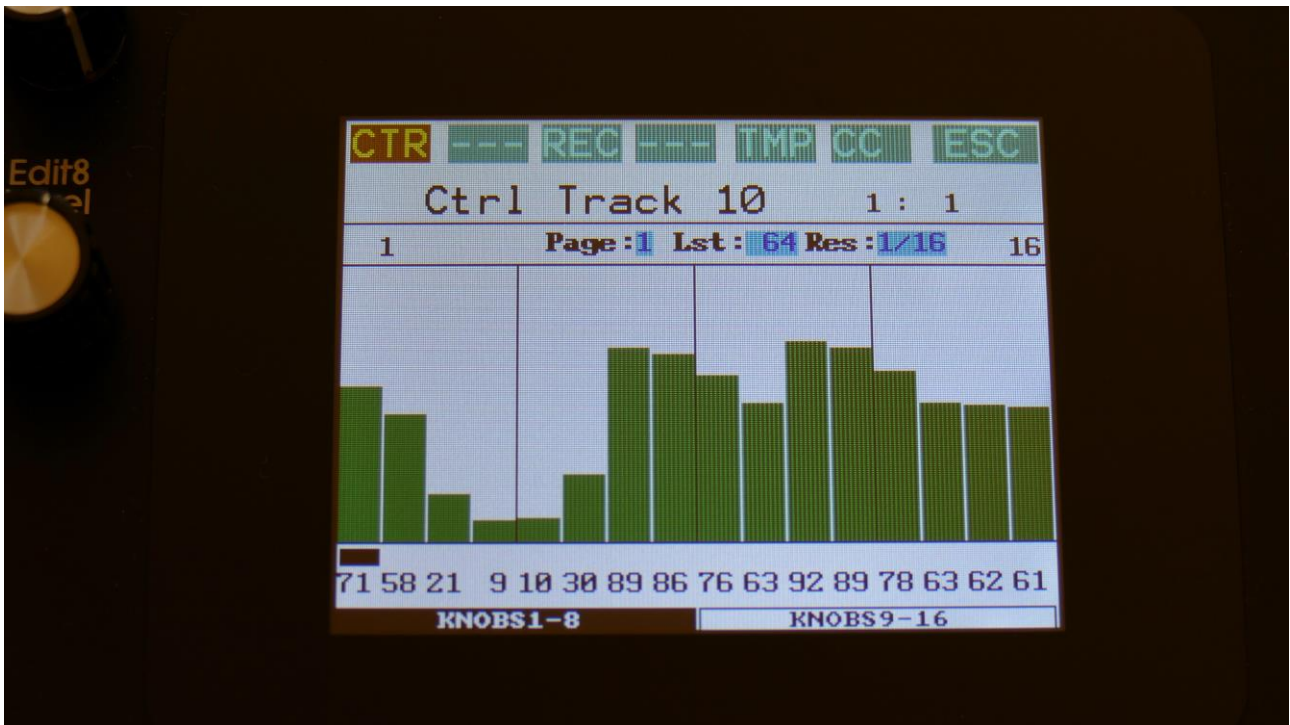
Via the controller tracks, it is possible to realtime record MIDI CC's, pitch bends and knob movements.

Templates are available, for quick modification of the controller track values.

Controller track Step Values

From the sequencer access page, touch Controller Track 1 to 16, to access these controller tracks, or touch Controller Track 17 to 32, to access these tracks. OR: Push and release the Func/Mute button, so that it lights up, and push step button 4/Ctrl Trk 1 time to enter tracks 1-16, and 3 times to enter tracks 17-32.

To access a specific track, push and hold the Steps/Part button, while pressing one of the step buttons.



This will be the first page, you will enter, showing the controller values, that will be outputted for each sequencer step, when the sequencer is playing back.

At most of this screen, you will see the value grid. Right above this, to the left and to the right, the first and last step numbers, of the steps that is currently shown in the value grid, are indicated.

Between these numbers, a few settable parameters is shown.

In the top left side, a small VU-meter is shown, that show the value currently being outputted by the track.

Here, when neither KNOBS1-8 or KNOBS9-16 are black:

Edit Knob 1: Has no function.

Edit Knob 2: Selects the bar to be viewed. 4 bars can be viewed, showing each 16 steps, for a total of 64 steps. The bar number is shown just above the values field (marked **Page**), and the first and the last step number viewed, is shown just above the values field, to the right and the left.

Edit Knob 3: Sets the last step that will be played back on this track, before it loops back, and plays the track again. Range: 1 to 128 steps. Value is shown as **Lst**.

The Last step can be set globally for all tracks at the Seq Main page.

Edit Knob 4: Sets the track resolution. Values: 1/64, 1/48, 1/32, 1/24, 1/16, 1/12, 1/8, 1/6, ¼, ½. Value is shown as **Res**.

Touching “ESC” will exit you to the Sequencer Access page.

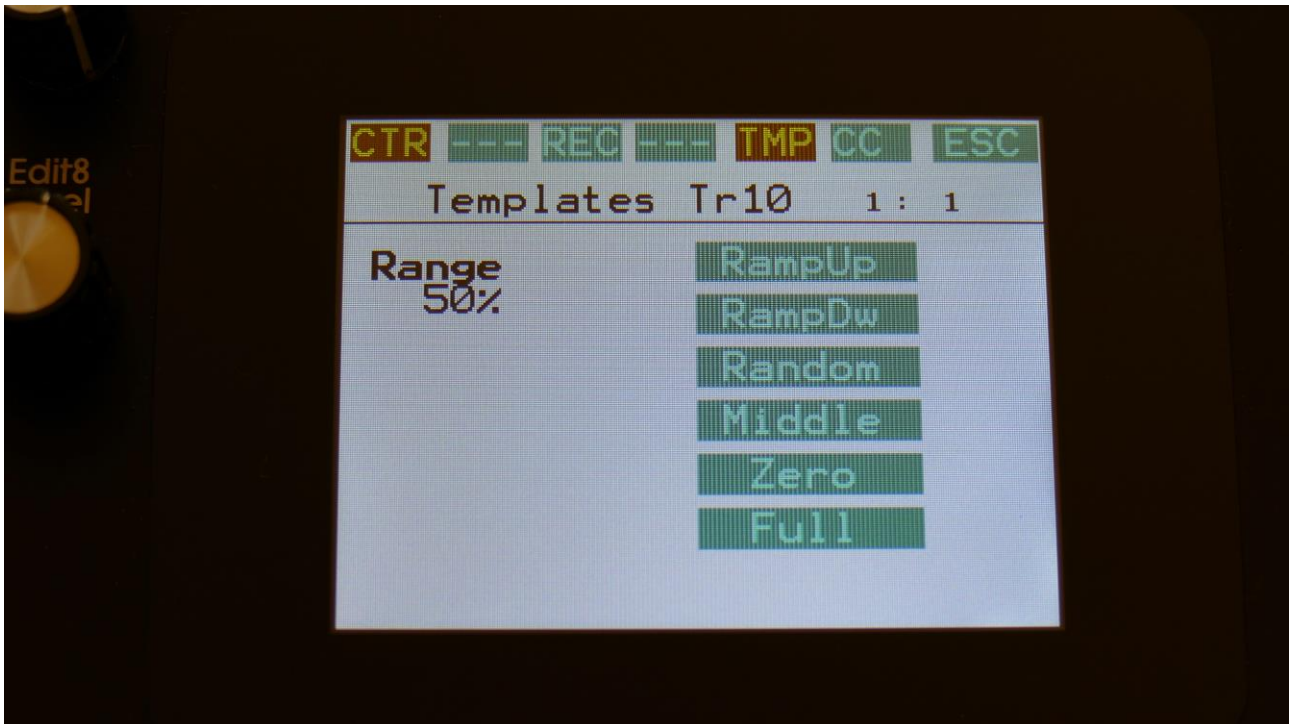
The controller values for each step is shown right below the value grid. If a black underscore is shown below the number, it means that the value are above 100. Controller values for each step can be changed, simply by touching the value grid.

Values can also be changed, using edit knob 1 to 8. In the bottom of the page, touch the field named “Knobs 1-8”, to edit the first 8 steps, using the edit knobs, or touch the field named “Knobs 9-16”, to edit the last 8 steps, using edit knob 1 to 8.

Controller Tracks Templates

Templates are available to make it easier to create ramps on a track and to randomize things. For the position track templ

When on either the Controller track step values page, touch TMP on the top navigation bar, to enter the templates page.



First turn Edit Knob 1, to set the range. This will set, how much the template routine will affect the values, that are already on the selected subtrack. When this is set to 100%, the template function will be absolute. Then touch any of the template boxes in the right side of the screen.

The templates work from step 1 to the last step that is set. Values after the last step are not affected.

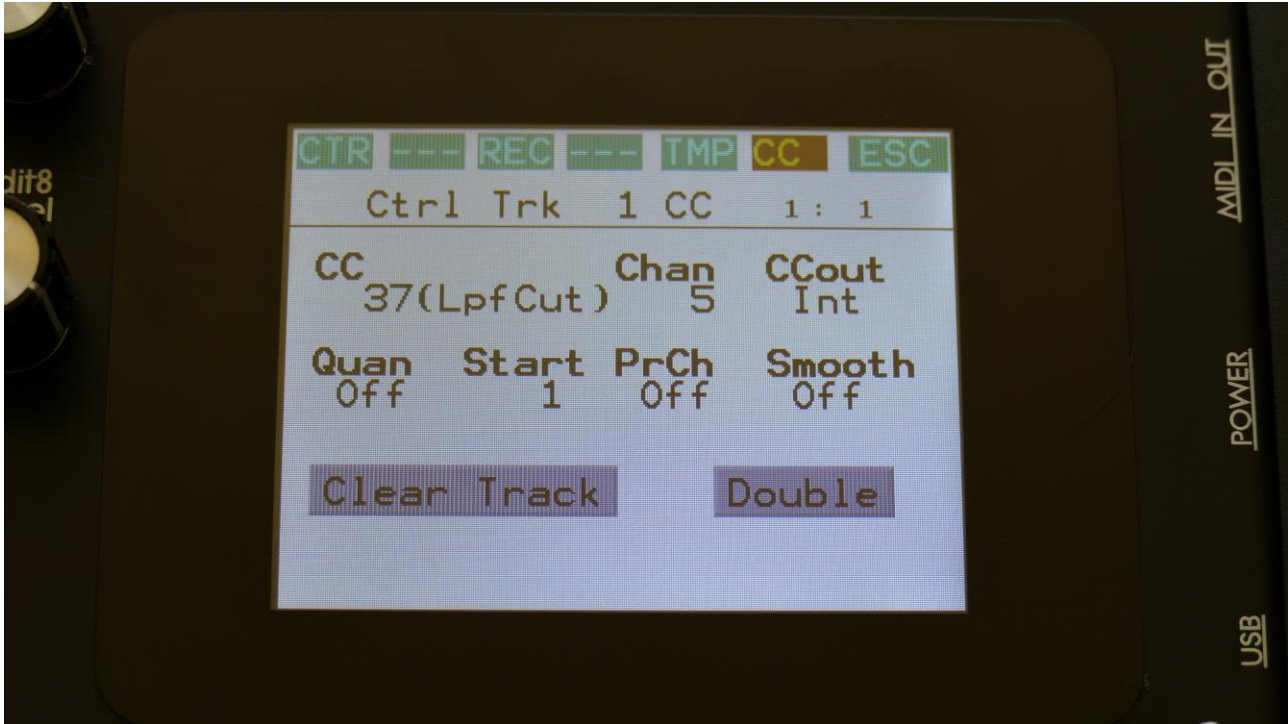
Available templates:

- RampUp**: Ramp Up. Will lower values on steps, that are placed before last step divided by 2, and raise values on steps, placed after last step divided by 2.
- RampDw**: Ramp Down. Will raise values on steps, that are placed before last step divided by 2, and lower values on steps, placed after last step divided by 2.
- Random**: Will randomize all steps between step 1 and last step, to a degree set by the range parameter.

- Middle:** Will move any values closer to the middle value.
- Zero:** Will lower any values.
- Full:** Will raise any values.

Controller Tracks CC page

Touch CC in the top navigation bar, to enter this page.



The parameters of this page:

CC: Off, 1(Morph), 2(SeqMrp), 3, 4(Knob1), 5(Knob2), 6, 7(Knob8), 8(Knob3), 9(Knob4), 10(Knob5), 11(Knob6), 12(Knob7), 13 to 15, 16 (Y-Mod), 17 to 122, CV IN 1 to 4, Pitch Bend. If the PccIN (Parameter CC's input) parameter on the MOR>COM2 page is turned on, it will also be possible to select many parameters for direct controller track control. The parameter names will be written in the parenthesis. Please see the list of parameter in the "MIDI Specs" section.

Setting this to any other value than "Off", will make the selected controller track output MIDI CC values or pitch bend values to MIDI out or internally, as set on the CCout parameter. When a number is set, this is the CC number that will be outputted. If any LD3 Edit Knob are assigned to this CC, the knob name will be written in parenthesis. If set to Pitch Bend, this is what will be outputted. The CC/Pitch Bend set here, are also the event that the track will realtime/step record. Controller track realtime and step recording are explained a little bit later.

Chan: 1 to 16. Sets the MIDI channel, that the selected MIDI CC will be transmitted to, when CCout is set to External, or the internal Synth Part number, that the CC's are sent to, when CCout is set to Internal.

CCout: Ext, Int. Sets if the controller track CC's/Pitch Bend's should be transmitted to an external MIDI device via MIDI Out (Ext), or to an internal Synth Part (Int). Regardless of this setting, the controller track are always available as internal modulation sources.

Quan: This can be set to either **Off**, **1v** or **1.2v**, for each controller track. When set to 1v, the output of the controller track is quantized to 1V per octave, when assigned as a source for the CV outputs. When set to 1.2v, the output is scaled to 1.2V per octave.

Start: 1 to 128. Sets the track start step. This will be the step, the track starts at, when hitting the Start/Stop button, or when entering this preset from another preset, while the sequencer is running.

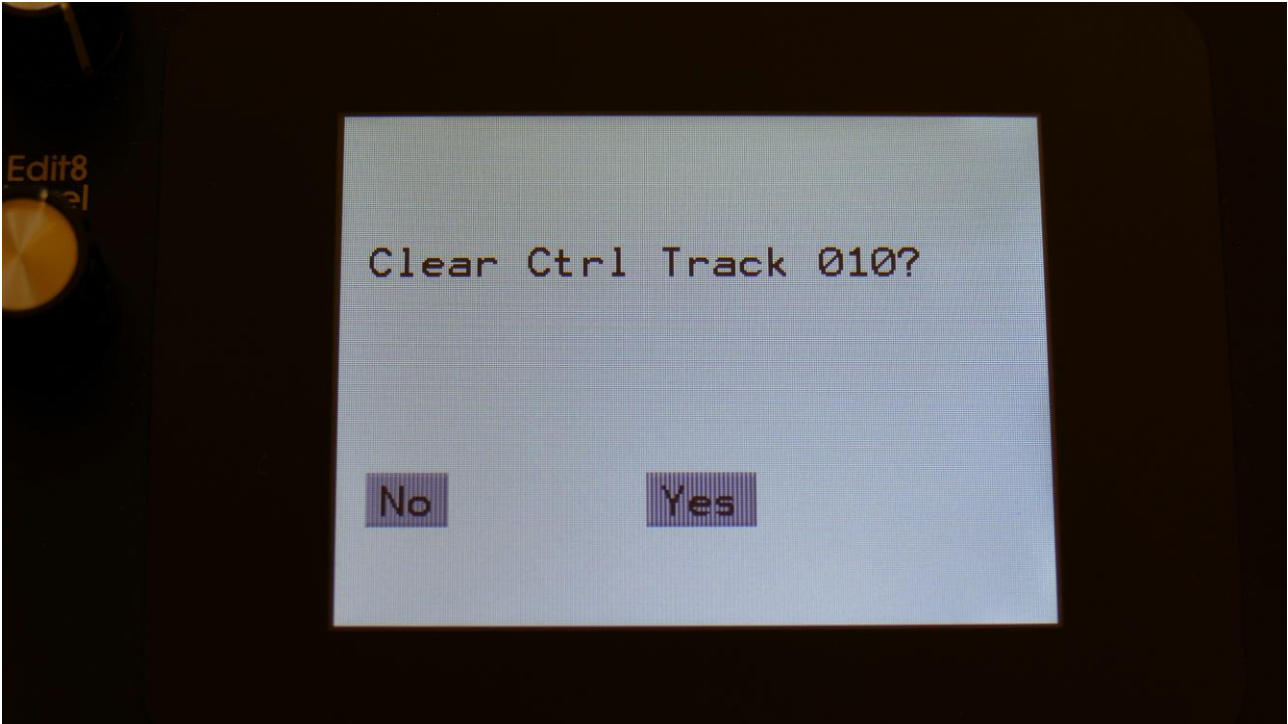
The Start step can be set globally for all tracks at the Seq Main page.

PrCh: Off, 1 to 128. When setting the PrCh parameter to any value between 1 and 128, a program change message will be transmitted on MIDI out, on the MIDI channel selected by the Chan parameter, every time this preset is selected. The CCout parameter has no effect on this. Each of the 32 controller tracks can transmit its own program change.

Smooth: Off, On. When this is set to On, extra controller values will be added between the controller value steps, to ensure smooth transition between these.

Clear Controller Track

From the track CC page, touch the **“Clear Track”** button, to enter this page:



Touching **“Yes”** from here, will clear all step on's of the currently selected controller track.

If you would not wish to do this, touch **“No”**, to exit to the track main page.

Double Controller Track

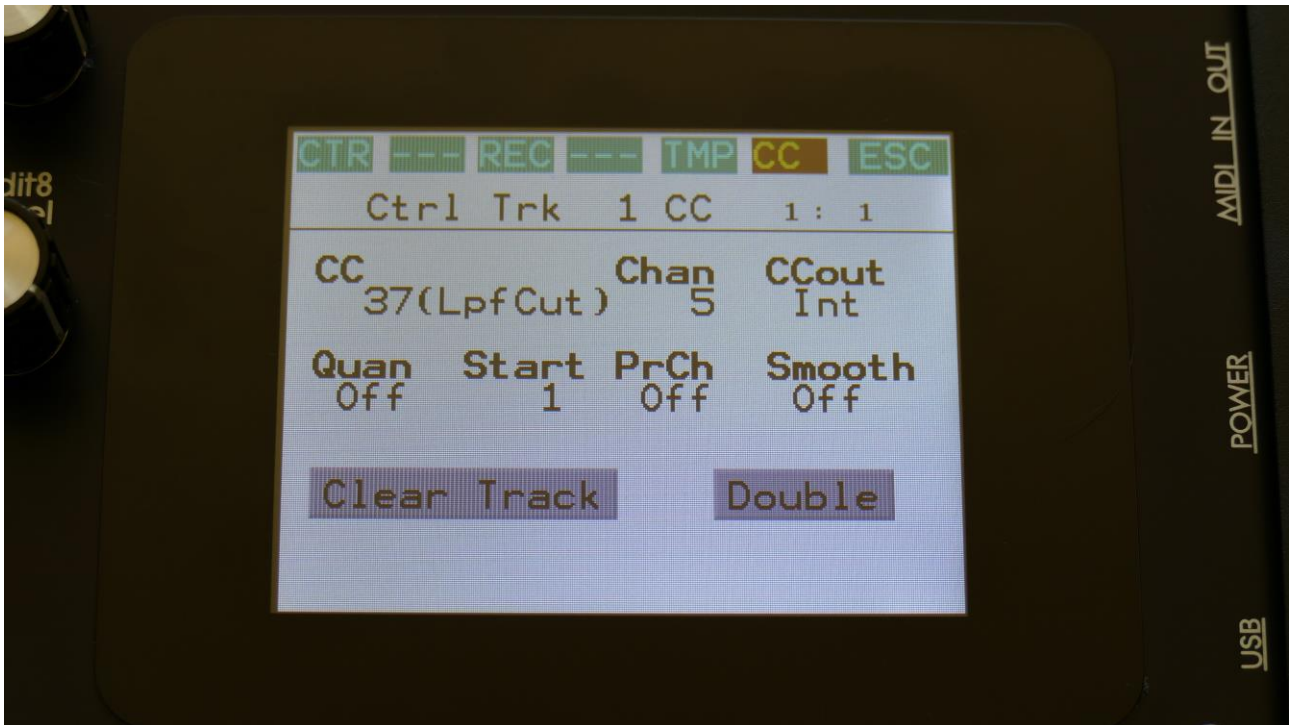
Touch the “Double” touch button, to double the length of the track. All events from the original track, is copied to the new half. Useful if you need some extra variation in a track.

PLEASE NOTE: The controller track lengths cannot exceed 128 steps.

Controller Tracks Realtime Recording

MIDI CC's, Pitch Bend, CV Inputs and knob movements can be realtime recorded on the 32 controller tracks.

To do this, on the controller track CC page, set it to the event type, that you would like to record, using the CC parameter.



The CC parameter can be set to any MIDI CC between 1 and 122.

In the top position, you will find CV In 1 to 4 and Pitch Bend, and at position 16 (CC16), you will find the touch screen keyboard Y-position.

The MIDI CC used for knob movement recording are named like this:

- 1(Morph)
- 2(SeqMrp)
- 4(Knob1)
- 5(Knob2)
- 7(Knob8)
- 8(Knob3)
- 9(Knob4)
- 10(Knob5)
- 11(Knob6)

12(Knob7)

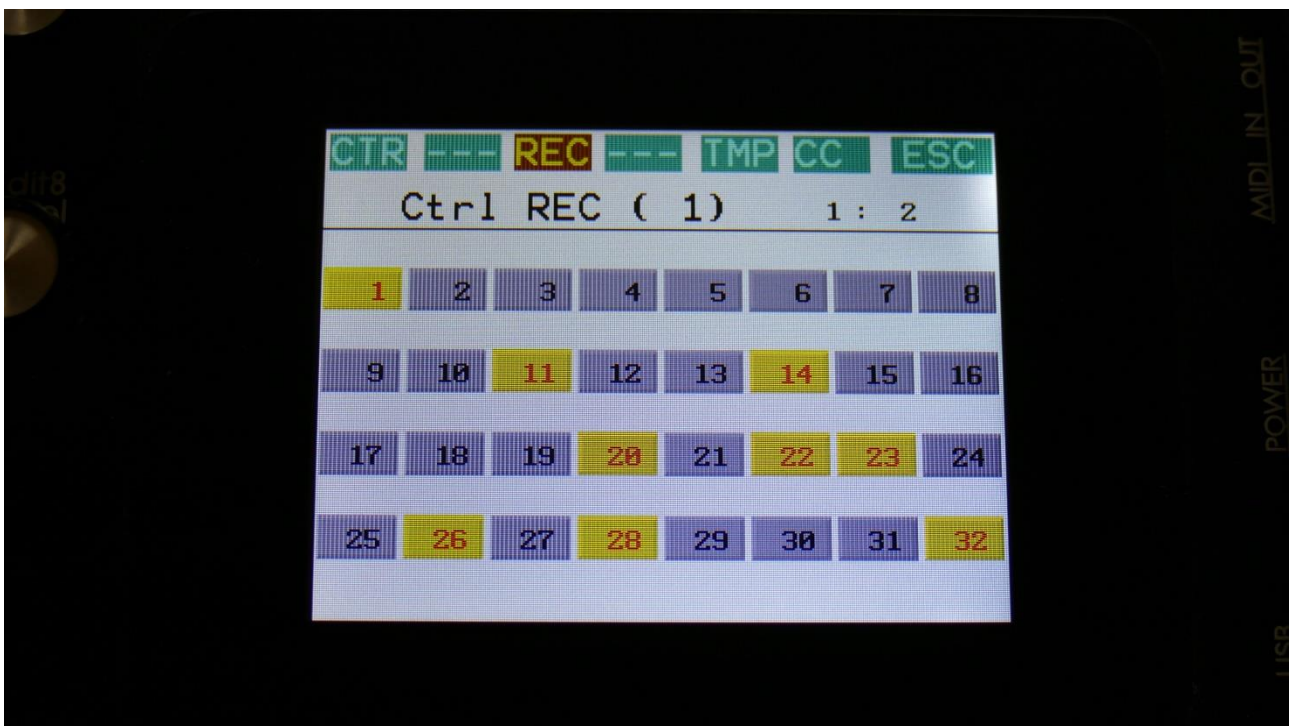
For knob movement recording, you should select one of these.

If the PccIN (Parameter CC's input) parameter on the MOR>COM2 page is turned on, it will also be possible to select many parameters for direct recording. The parameter names will be written in the parenthesis. Please see the list of parameter in the "MIDI Specs" section.

Set CCout to Int (internal) or Ext (external), depending on, if you like the controller track to control LD3 itself, or some external MIDI gear.

Set the track to the desired MIDI channel, using the Chan parameter. If the track is set to internal operation, the channel number is the part number, that the track will output to.

Touch REC on the top navigation bar, to enter this page:



Select the controller track(s), that you would like to realtime record, by touching the square, that has the same number(s) as the track(s), so that these turns yellow. Now they are armed.

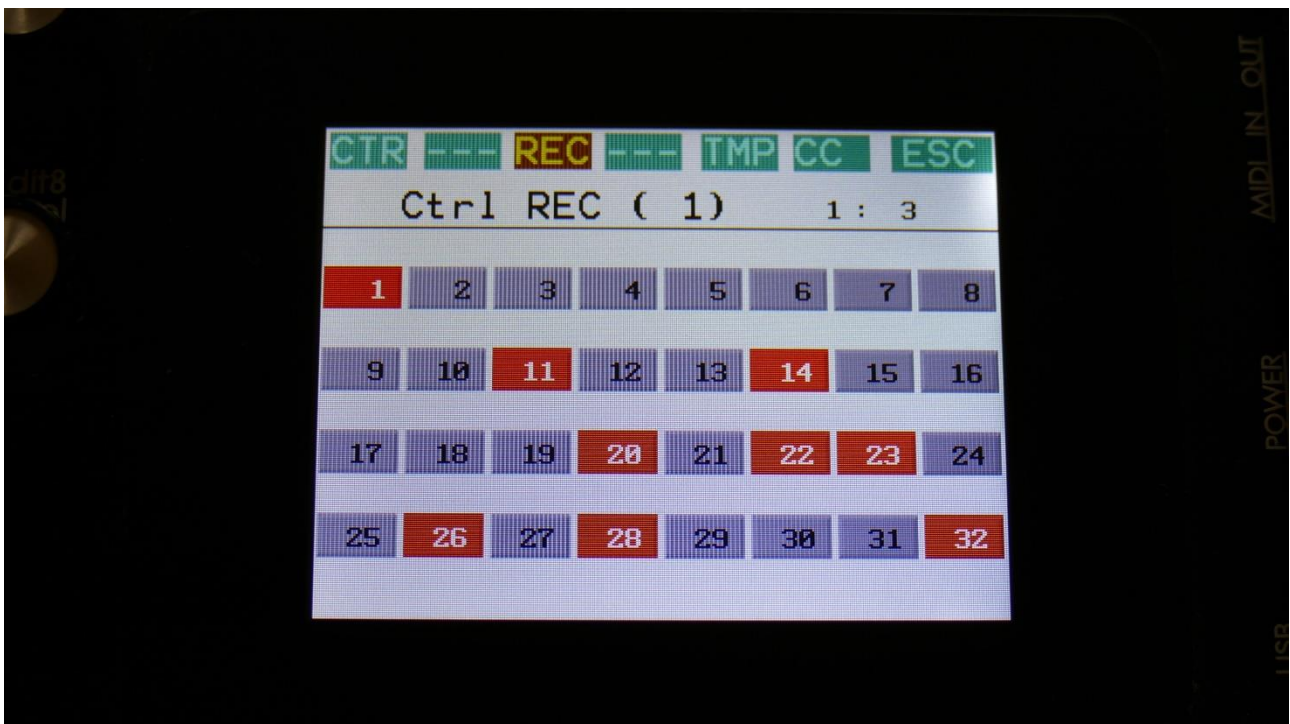
Set the length of the controller tracks(s), if desired.

Now set LD3 in sequencer recording mode, by touching the REC field on the sequencer main page or on the preset select page, so that it turns red, or push and hold the Func/Mute button, while pressing the Start/Stop button.

Start the sequencer. Tweak the knobs and/or MIDI CC's.

The controller tracks, that are set in record mode, will record until the track reaches the last step, and starts over again. Then recording will switch off.

As soon as the controller tracks are recording, the squares of the recording tracks will turn red.



If the sequencer was running, when you switched the controller tracks into recording mode, they will stay armed (yellow), until the controller track passes step 1. Then they will switch to recording mode, and the squares will turn red, to indicate that.

When a controller track reaches its last step, recording will stop, and its square will turn grey.

Recording parameter tweaks to a controller track

First, set up the controller track, as described on the previous pages of this manual.

Stop the sequencer, if it is running.

Touch REC in the upper menu bar. Activate recording for the selected controller track.

Go to the CTR page, and set the controller track length.

Now, go to the Synth section page and select the part, where the parameter is located.

Put Spazeboard6 in realtime recording mode, by holding down the Part/Mute button, while pressing the Start/Stop button.

Press the Start/Stop button, to start the sequencer.

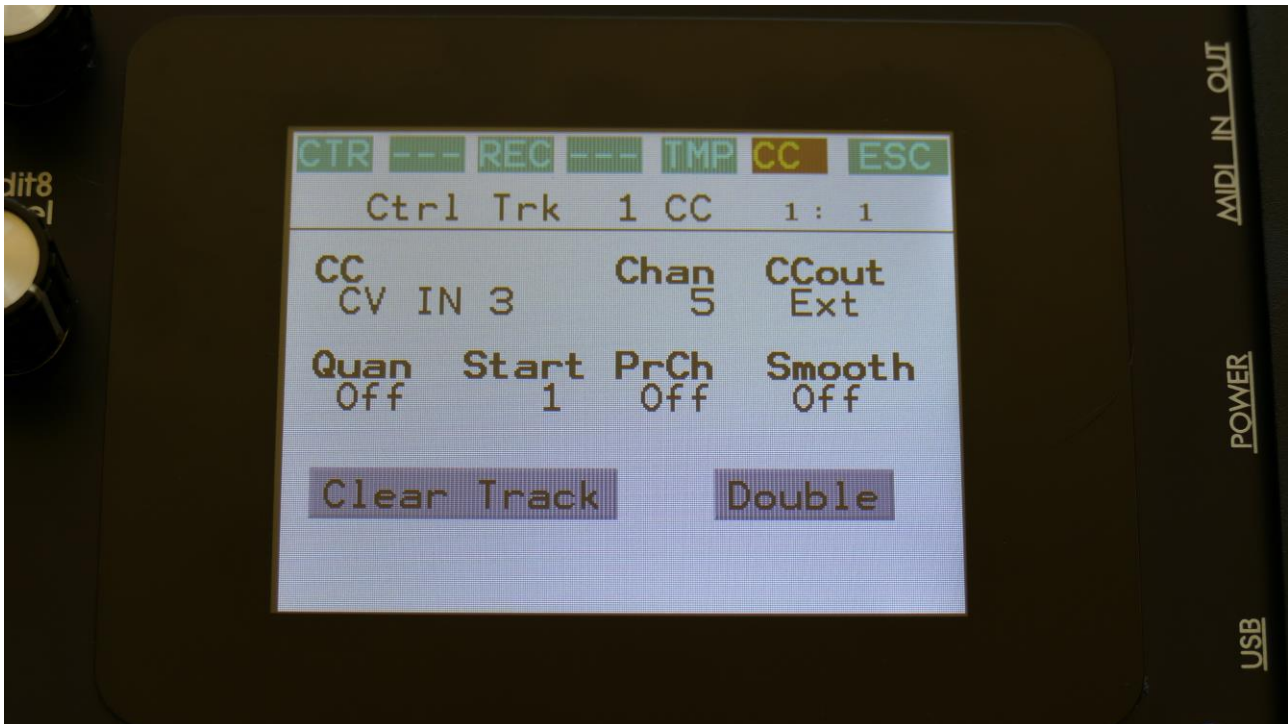
It will now count in for 2 bars, and then start to record. Tweak the parameter.

When the controller track reaches its end step, recording will stop, and the controller track will start to play back. You will now see the parameter value move by itself!

CV Inputs Recording

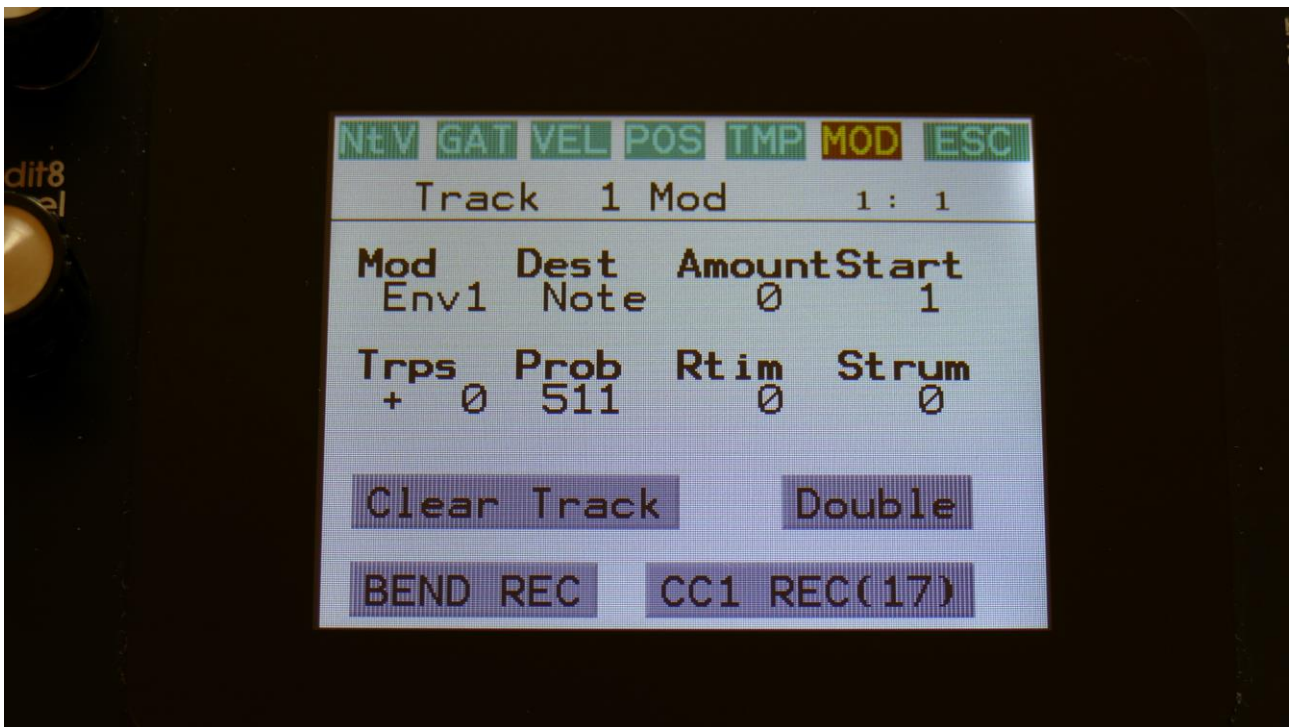
The 4 (optional) CV inputs can also be recorded to the sequencer controller tracks, just like MIDI CC's and knob movements.

To do this, simply select CV In 1 to 4 to be the CC on a controller track, and follow the procedure for controller track realtime recording.



You will, most probably, like to switch the Smooth parameter On, to get smooth transitions between the recorded CV values.

Pitch Bend and CC recording auto setup



From the sequencer note track Mod pages, it is also possible to make LD3 automatically set up controller tracks to realtime record pitch bend and CC events.

When using this function, the controller track is set to the double resolution as the note track, and therefore also the double length, to make the tracks match.

Note track 1 will set up controller track 1 for pitch bend recording and controller track 17 for CC recording.

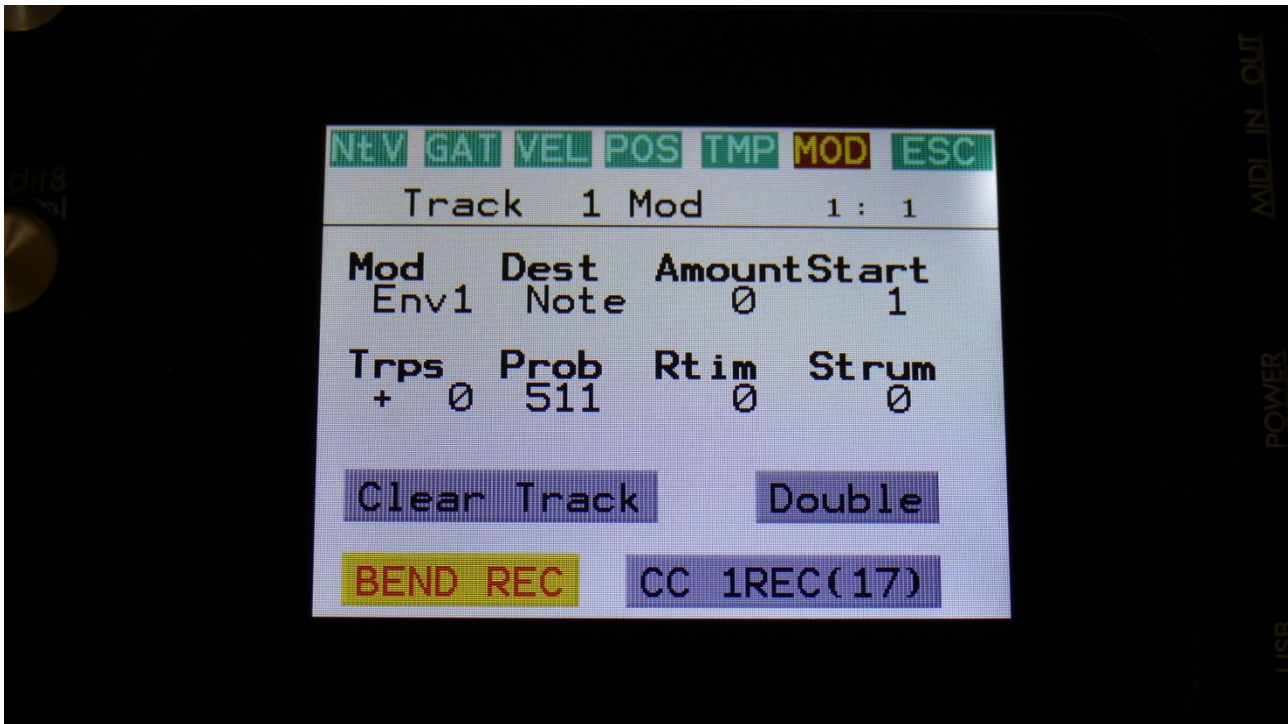
Note track 2 will set up controller track 2 for pitch bend recording and controller track 18 for CC recording.

-
-
-

Note track 16 will set up controller track 16 for pitch bend recording and controller track 32 for CC recording.

The CC number to be recorded, will be the CC number, that the controller track is already set to. If the controller track is not already set to a CC number, CC1 will be recorded.

To initiate Pitch Bend recording



- Make sure that sequencer realtime recording is switched off.
- Set the desired length and resolution of the note track.
- Go to the Mod page, and touch BEND REC, so that it turns yellow.
- Start realtime recording.
- If the sequencer is already running, recording will start, when the controller track passes step 1. The Bend Rec touch button will then turn red, to indicate that recording is active.
- Pitch Bend messages will only be recorded the first time, the controller track is running. When the controller track passes its last step, recording will be switched off, and the Bend Rec touch button will turn grey.

Follow the same procedure, to record CC events, plus set the desired CC number on the controller track.

Controller Tracks Step Recording

It is also possible to step record controller events.

To initiate step recording:

- Set the controller track in realtime recording mode, as described in a previous chapter.
- Start the sequencer playback, if desired. Step recording is possible no matter if the sequencer is running or not.
- Push and release the Steps/Part button, so that it lights up, and puts the 16 step buttons in step mode.
- Push the step button, that has the position for the step, that you would like to record. This will now flash.
- An armed controller track rec indicator (yellow), will now switch to recording mode (red).
- Set the Knob/MIDI Controller to the value, you would like this step to hold.
- Select another step to record.

Synth and Sequencer Morphing

On LD3 it is possible to morph between 2 layers of parameters, layer A and B, by turning the Morph knob for the synth parameters, and the Seq Morph knob for the sequencer parameters. The Morph knob transmits and receives MIDI CC 1 (The Mod Wheel) via MIDI in and out, and the Seq Morph knob transmits and receives MIDI CC2.

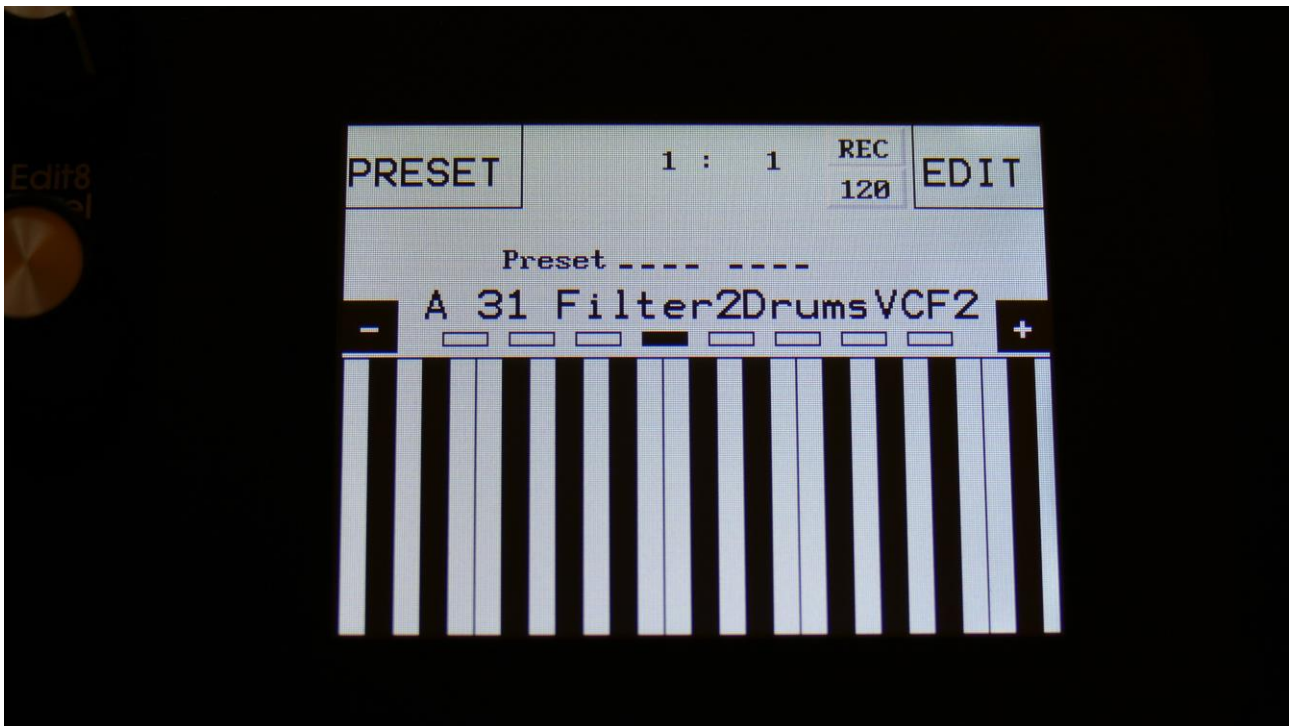
To access the parameters of the second layers, on any synth and sequencer page, hit the Morph Set button, so it lights up. Now the parameters of layer B will be shown on the screen, and you can edit these, using Edit Knob 1 to 8, just like the layer A parameters.

All continuously adjustable parameters can be morphed. Switches are not morphed, except for the note tracks steps on/off.

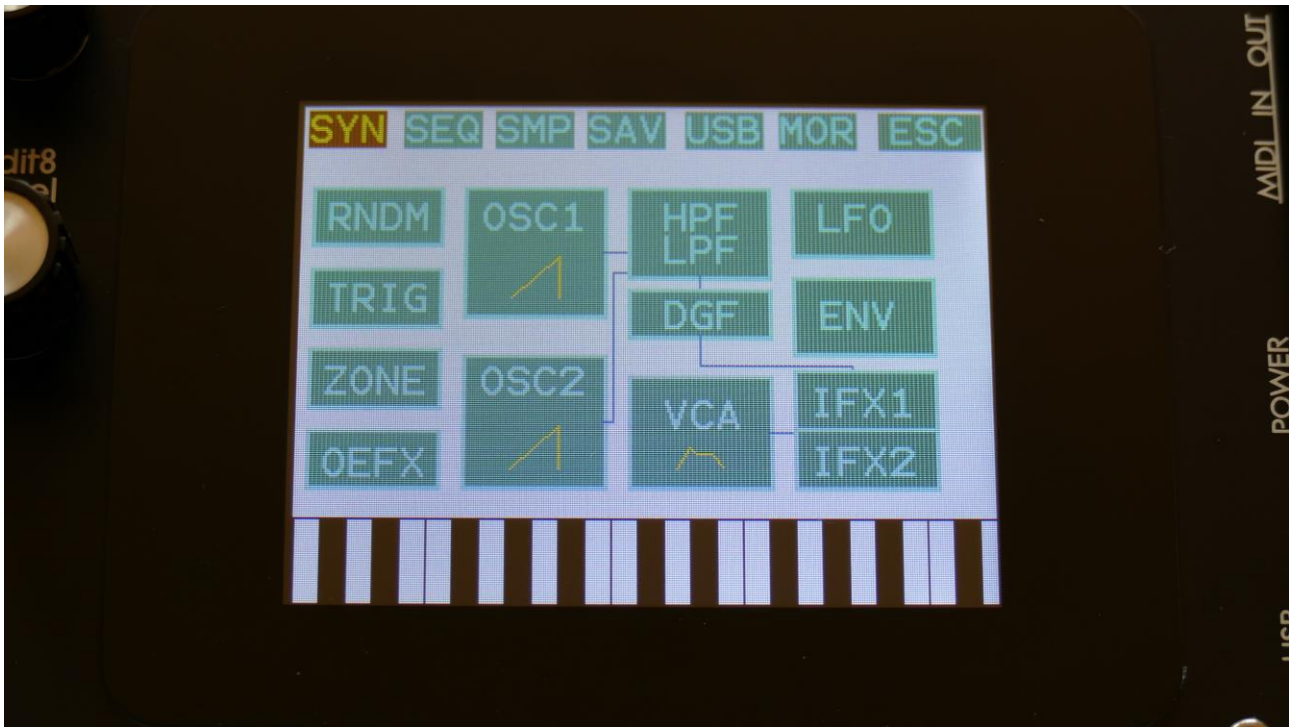
In the MOR section under MORPH, it is possible to set static Morph knob values, and to modulate the morph knobs.

All morph settings are stored within each Preset. Remember to save the Preset, to not loose any great sounds.

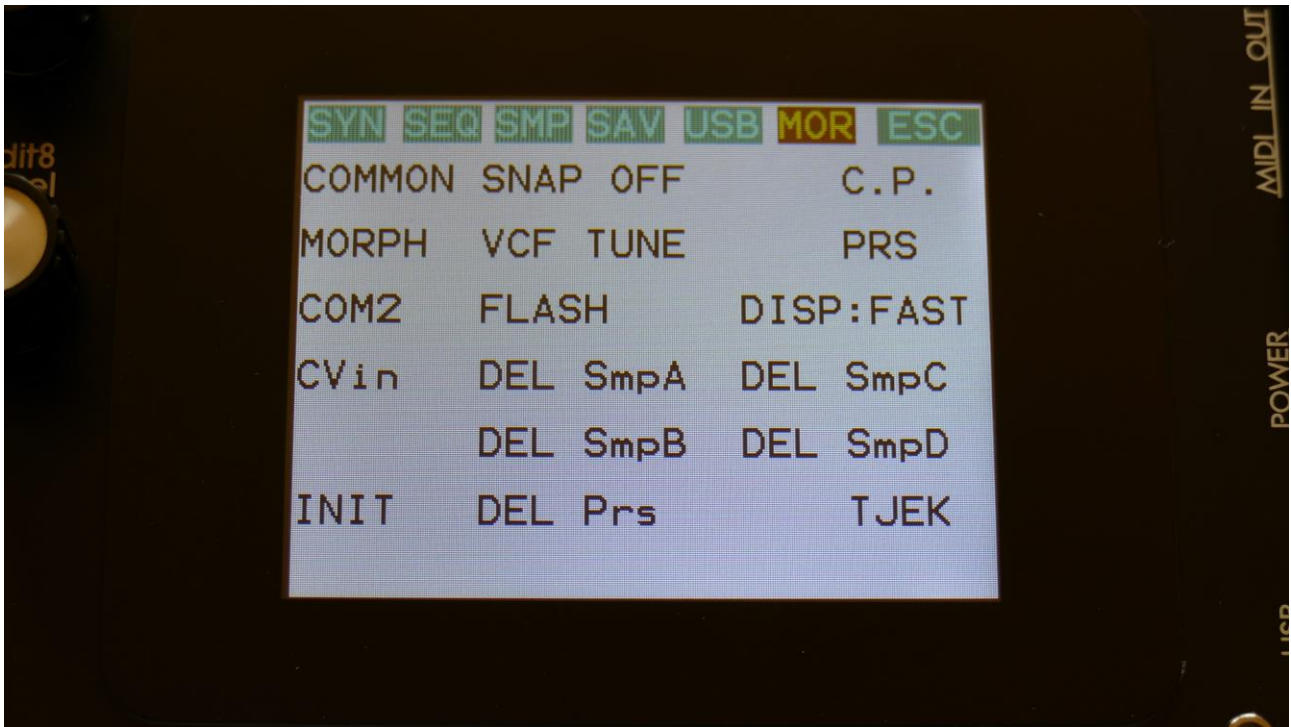
More... preset parameters/Setup



From the Preset Select page, touch the Edit field.



On the navigation bar at the top of the screen, touch MOR.



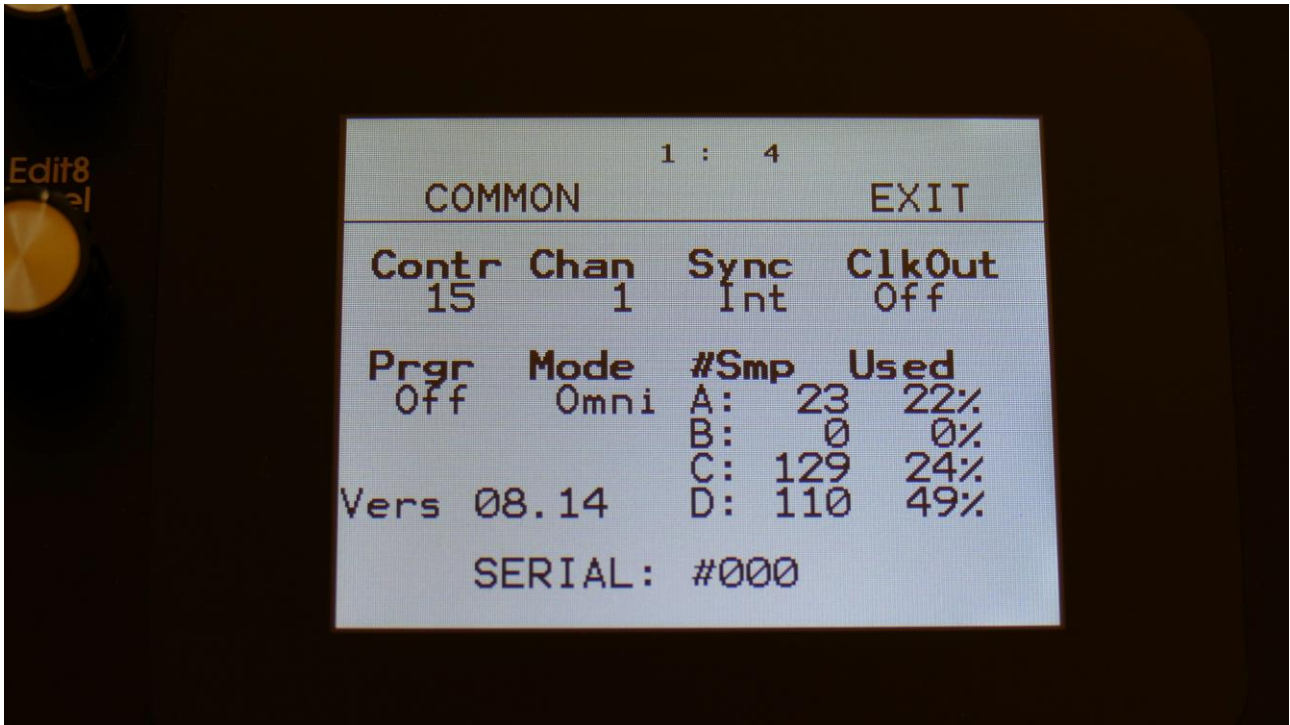
This will get you to the More... page. Here you can set up all the basic LD3 things, like global MIDI channel, morphing and CV in's, tune the analog filters, check and erase the LD3 memory, initialize the preset and change to song mode.

On the following pages, each section of the More... page is described.

Common Settings

On this page, you can adjust the display contrast and MIDI settings and check the sample memory.

From More... page, touch COMMON to enter this page:



The following parameters can be adjusted on this page, using edit knob 1 to 8:

Contr: Adjusts the display contrast. This setting is automatically saved, and will be remembered after a power off.

Chan: The LD3 global MIDI channel. All MIDI data to LD3 will be received on this channel, if the Mode parameter is set to Sel (Explained in a moment). MIDI CC's from the Edit/Quick Edit Knobs and the Morph knob will be transmitted on this channel, if the Mode parameter is set to Sel. This setting is automatically saved, and will be remembered after a power off.

Sync: Sequencer synchronization. This setting is automatically saved, and will be remembered after a power off. Settings:

-**Int:** LD3's sequencer will be clocked internally at the rate set by the "Tempo" parameter in each preset.

-**Ext:** LD3's sequencer will be clocked externally, by MIDI clock events applied to MIDI in. Sequencer play and stop can also be controlled externally.

ClkOut: Settings: Off, On. Selects whether or not LD3 will re-transmit MIDI clock and start/stop events to MIDI out. Even if LD3 is syncing to an external MIDI clock source, it will re-transmit this, if this parameter is on. This setting is automatically saved, and will be remembered after a power off.

Prgr: On/Off. When switched on, LD3 will receive MIDI program and bank changes. Bank change is MIDI CC 32.

Mode: LD3 MIDI input mode. This setting is automatically saved, and will be remembered after a power off. Can be set to:

-**Sel:** The selected synth part is controlled on the LD3 base MIDI channel.

-**Omni:** The selected synth part is controlled on any MIDI channel.

-**OmTh:** Omni Thru. The selected synth part is controlled on any MIDI channel. MIDI events are channelized to MIDI out on the MIDI channel selected on the Synth Part TRIG page.

-**Mult:** Multi timbral. All 16 parts works in multi timbral mode, and is controlled by the MIDI channel/key range set on each parts ZONE page.

-**MuTh:** Multi timbral/Thru. All 16 parts works in multi timbral mode, and is controlled by the MIDI channel/key range set on each parts ZONE page. MIDI events are channelized to MIDI out on the MIDI channel selected on the Synth Part TRIG page.

#Smp: Shows how many samples that are held in sample bank A, B, C and D.

#Used: Shows in percent how much of the sample memory of each sample bank is occupied by samplings.

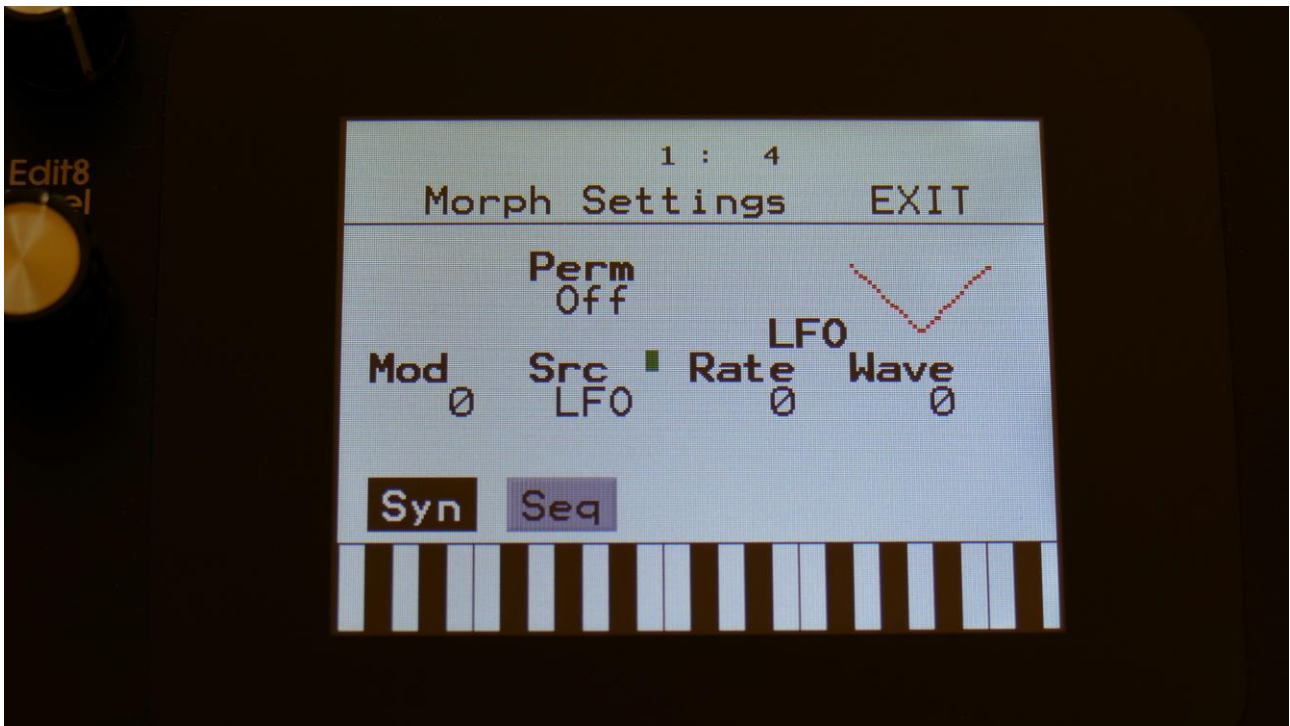
In the bottom of this screen, the currently installed firmware version number is shown, and the serial number for your LD3.

Morph Setup

On the Morph Setup page, it is possible to set a permanent morph value, that will override the Morph knob, if you wish to always have a preset playing back in a certain morph position.

It is also possible to modulate the morph knobs, using a limited number of modulation sources, including a morph LFO.

From More... page, touch MORPH to enter this page:



This is the settings for the Morph knob, that affect the Synth Parts, LFO's and effect processors. Touch the Syn touch button, to access this page.

Parameters on this page:

Perm: Morph knob permanent value. Will override the value of the Morph knob, and keep morph on the permanent value plus modulation. Range: Off, 1 to 511.

Mod: Morph knob modulation amount. 0 to 511.

Src: Morph modulation source. Sources are:

-**LFO:** The morph section LFO. Described below.

-**Seq1 to Seq8:** Sequencer controller tracks 1 to 8.

-**CV1 to CV4:** CV inputs 1 to 4.

-**Kybd:** The last note number value received

-**Velo:** The last note velocity value received

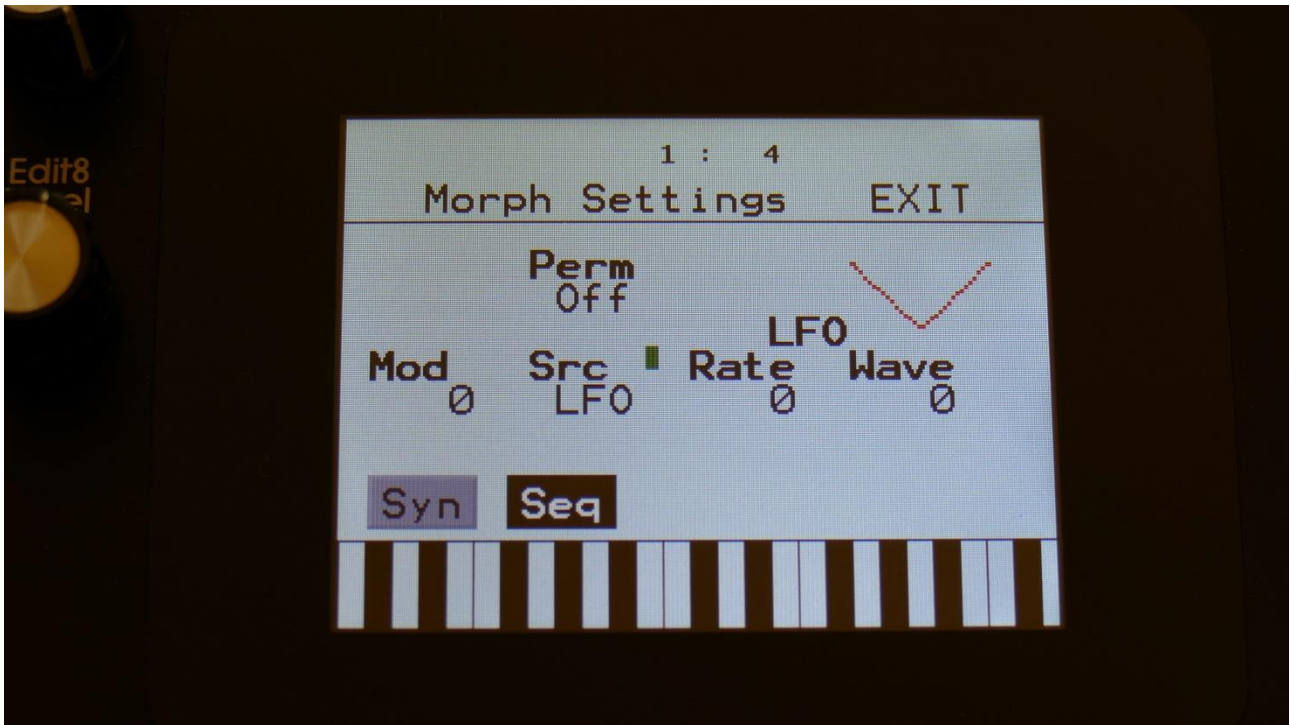
-**TouY:** Touch screen keyboard Y position.

Rate: Morph LFO rate. Sets the tempo of the morph section LFO, that can only modulate the morph parameter.

Wave: Morph LFO output waveform. Morphs between triangle, sawtooth, square, pulse and FM waveforms. The FM waveforms are high frequency waves, that can be used for FM synthesis.

Sequencer Morph Knob Setup

Touch the Seq touch button, to enter this page.



This is the settings for the Seq Morph knob, that affect the Sequencer.

Parameters on this page:

Perm: Seq Morph knob permanent value. Will override the value of the Seq Morph knob, and keep morph on the permanent value plus modulation. Range: Off, 1 to 511.

Mod: Seq Morph knob modulation amount. 0 to 511.

Src: Seq Morph modulation source. Sources are:

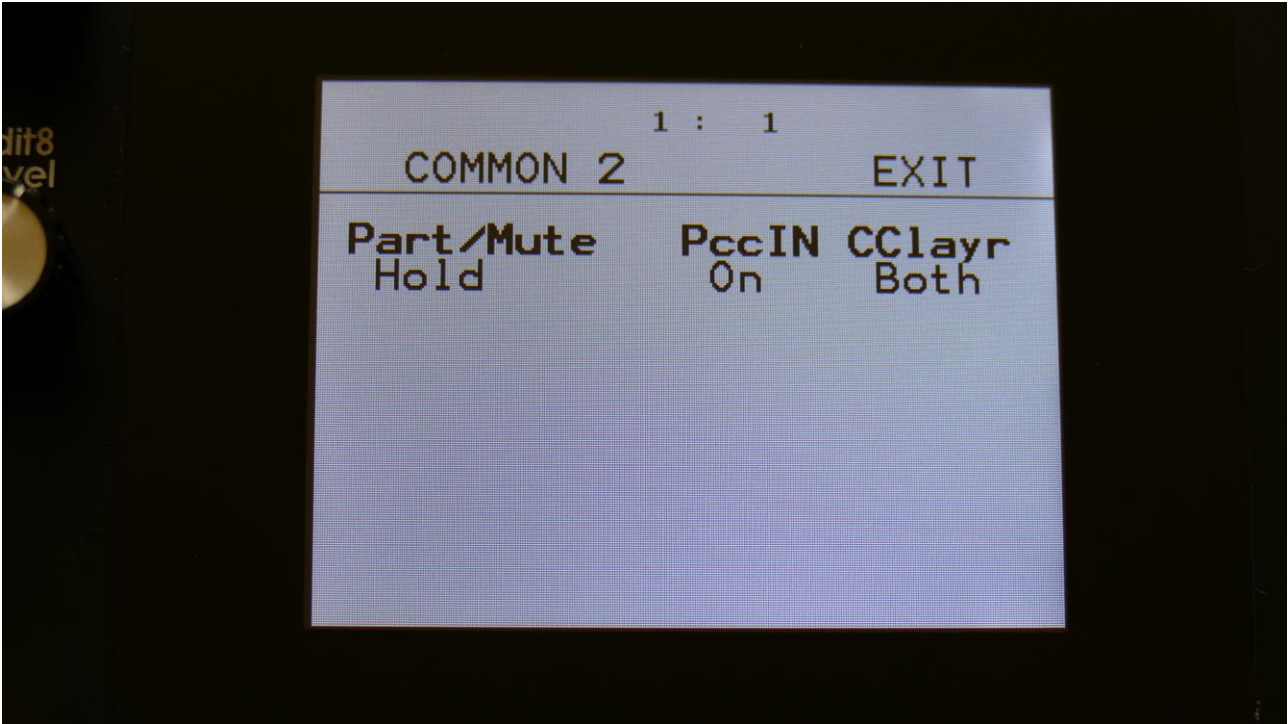
- LFO:** The morph section LFO. Described below.
- Seq1 to Seq8:** Sequencer controller tracks 1 to 8.
- CV1 to CV4:** CV inputs 1 to 4.
- Kybd:** The last note number value received
- Velo:** The last note velocity value received
- TouY:** Touch screen keyboard Y position.

Rate: Seq Morph LFO rate. Sets the tempo of the morph section LFO, that can only modulate the morph parameter.

Wave: Seq Morph LFO output waveform. Morphs between triangle, sawtooth, square, pulse and FM waveforms. The FM waveforms are high frequency waves, that can be used for FM synthesis.

Common Settings 2 (COM2)

On this page you can set if the part select and part mute buttons should latch or not, and if Spazeboard6 should receive parameter MIDI CC's.



Here you can, by rotating Edit Knob 1, set the **Part/Mute** buttons to:

- Hold:** You will have to hold the Func/Mute or Steps/Part buttons, while pushing a step button, to select or mute a part.
- Latch:** When pushing and releasing either the Func/Mute or the Steps/Part buttons 2 times, the button will start to flash, and you can select or mute parts, by pushing the step buttons. It is still possible to select and mute parts, by holding any of these buttons.

PccIN: Off, On. Parameter CC Input.

If this parameter is set to On, many of the Spazeboard6 parameters will be hard-assigned to MIDI CC numbers. This also makes it possible to control the same parameters directly from the sequencer controller tracks, and to realtime record parameters tweaks on the controller tracks. For a complete list of controllable parameters, please see the "MIDI Specs" section of this manual.

CClayr: CC control to Morph Layer. Sel, Both.

Sel: Only the Morph layer selected by the Morph Set button, will be affected by MIDI in and Sequencer CC control.

Both: Both Morph layers will be affected by MIDI in and Sequencer CC control, at the same time.

CV Inputs (optional)

With the optional analog board and CV inputs/outputs board installed, LD3 gets 4 CV inputs. These can be selected as modulation sources, for any parameter that can be modulated.

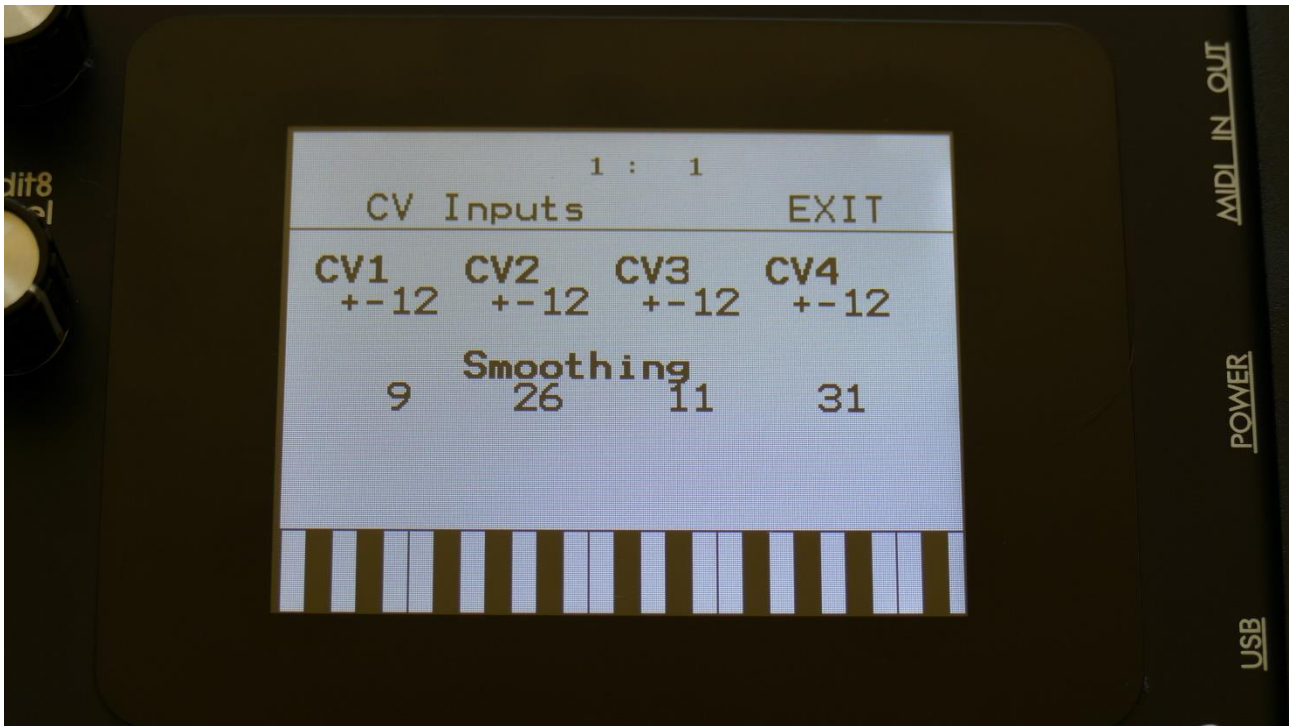
LD3 accepts CV voltages of up to +/- 15 volts. The voltage range of each CV input can be set up to a number of different configurations.

It is also possible to set up each CV input as a trigger for any of the 16 synth parts of LD3, and to set them up as clock sources, start/stop and reset control for the sequencer.

By setting a CV input up to trigger a part, and using another CV input to modulate the Part Trigger Note, CV/Gate to MIDI is possible.

Remember to save all edits you do to the CV Inputs. Else they will be lost when you change preset, or turn LD3 off. See how to in the "Save Preset" section.

From More... page, touch CVin to enter this page:



Here you can, by turning Edit Knob 1 to 4, set up each CV input. The possibilities for each input are:

+/-12: The CV input will work in the range of +/- 12 volts –standard eurorack voltages.

+12: The CV input will work in the range of 0 to 12 volts.

+/-5: The CV input will work in the range of +/- 5 volts.

+5: The CV input will work in the range of 0 to 5 volts.

Trg1 to Trg16: The CV input will work as a trigger input for Synth Part 1 to 16. As a modulation source, it will work in the +/- 12 volts range.

1/32: For each pulse it receives, the sequencer will increment 1/32 step. The internal sequencer clock, and any MIDI clock will be ignored. MIDI clock will still be transmitted, if the ClkOut parameter in the COMMON menu is set to on.

1/24: For each pulse it receives, the sequencer will increment 1/24 step. The internal sequencer clock, and any MIDI clock will be ignored. MIDI clock will still be transmitted, if the ClkOut parameter in the COMMON menu is set to on.

1/16: For each pulse it receives, the sequencer will increment 1/16 step. The internal sequencer clock, and any MIDI clock will be ignored. MIDI clock will still be transmitted, if the ClkOut parameter in the COMMON menu is set to on.

1/12: For each pulse it receives, the sequencer will increment 1/12 step. The internal sequencer clock, and any MIDI clock will be ignored. MIDI clock will still be transmitted, if the ClkOut parameter in the COMMON menu is set to on.

1/8: For each pulse it receives, the sequencer will increment 1/8 step. The internal sequencer clock, and any MIDI clock will be ignored. MIDI clock will still be transmitted, if the ClkOut parameter in the COMMON menu is set to on.

1/6: For each pulse it receives, the sequencer will increment 1/6 step. The internal sequencer clock, and any MIDI clock will be ignored. MIDI clock will still be transmitted, if the ClkOut parameter in the COMMON menu is set to on.

s/s: Will make the sequencer start or stop every time it receives a pulse or a gate on that input.

Rst: Will make all tracks of the sequencer reset to their respective start step every time a pulse or a gate is received on that input.

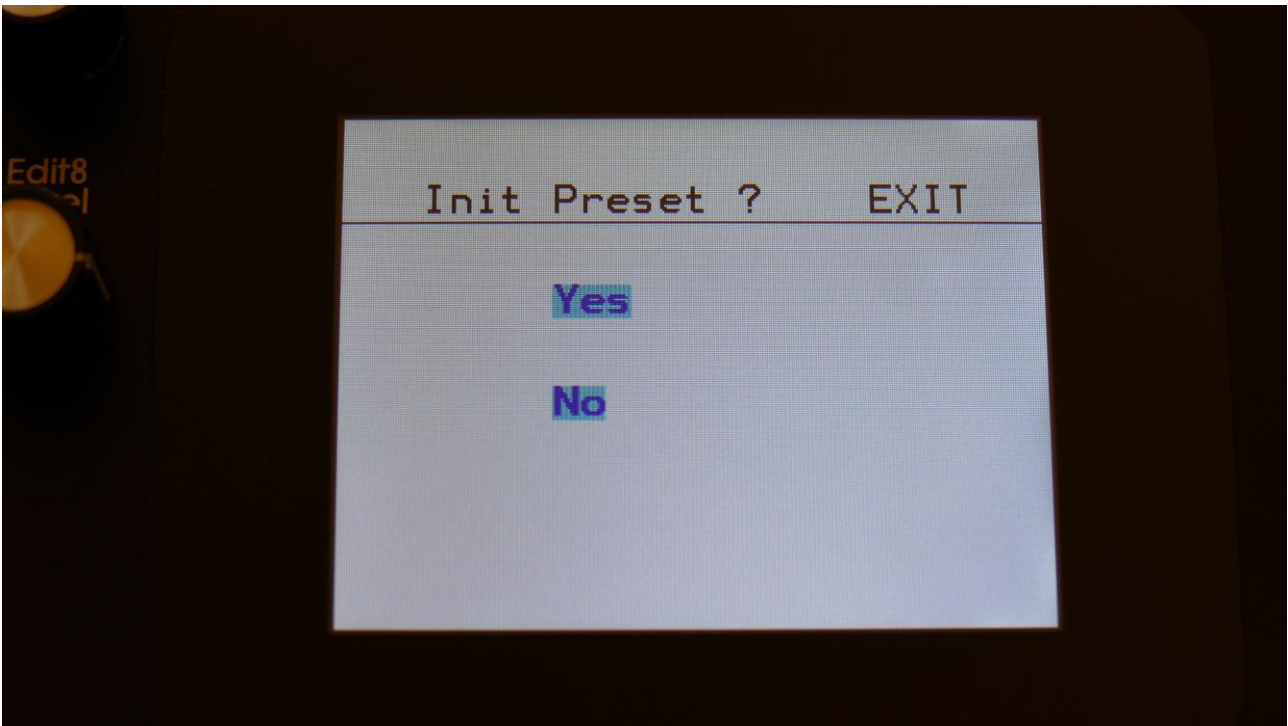
Smoothing: If the CV inputs sounds unstable, they can be smoothed out by turning up this smoothing parameter.

Initialize Preset

If you would like to start out from scratch, with a completely clean preset, this is the function to use.

Please notice that the preset on the selected location are not immediately wiped out. It will not be overwritten until you save the new preset, so even though you have initialized it, it can still be recalled, if you should regret, by selecting another preset, and then select this one again, or by using the reload function.

From More... page, touch INIT to enter this page:

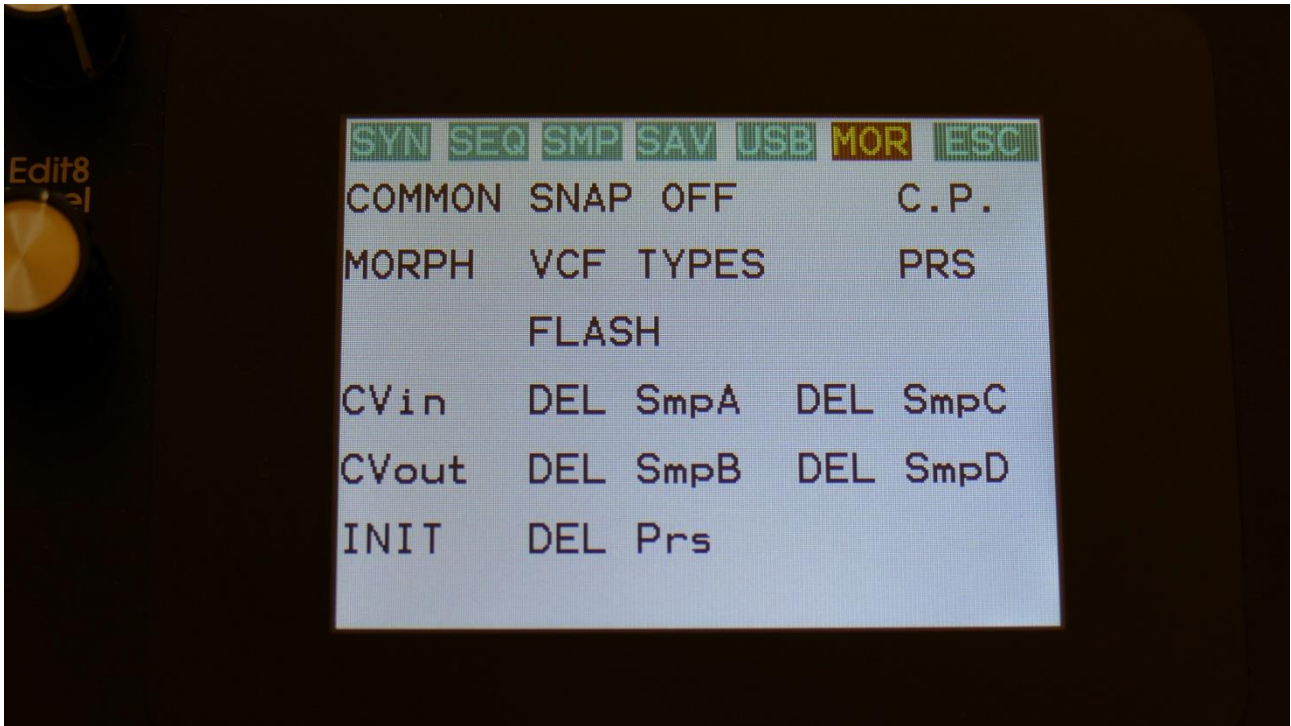


LD3 will now ask you to confirm.

Press "Yes" to initialize the currently selected preset and return to the previous page, or press "No" just to exit, without initializing.

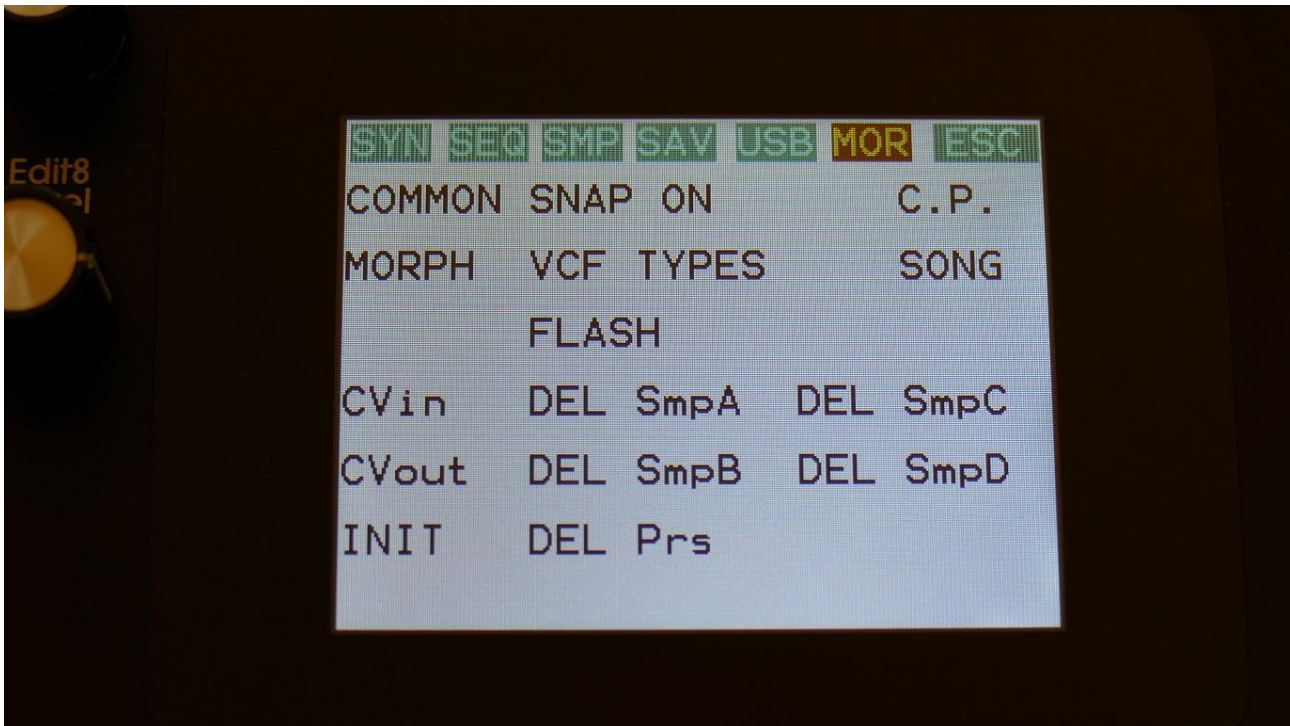
Parameter Snap Mode

A synth part parameters “Snap” mode are available. When this is set to “On”, any synth parameter value will not be changed, when turning a knob, until it passes the original value, that the parameter was set to. This will avoid “jumps” in the sound.



On the MOR page, touch the “SNAP OFF” text.

Parameter snap mode will now be turned on:



To turn it off, touch it again.

VCF Tune

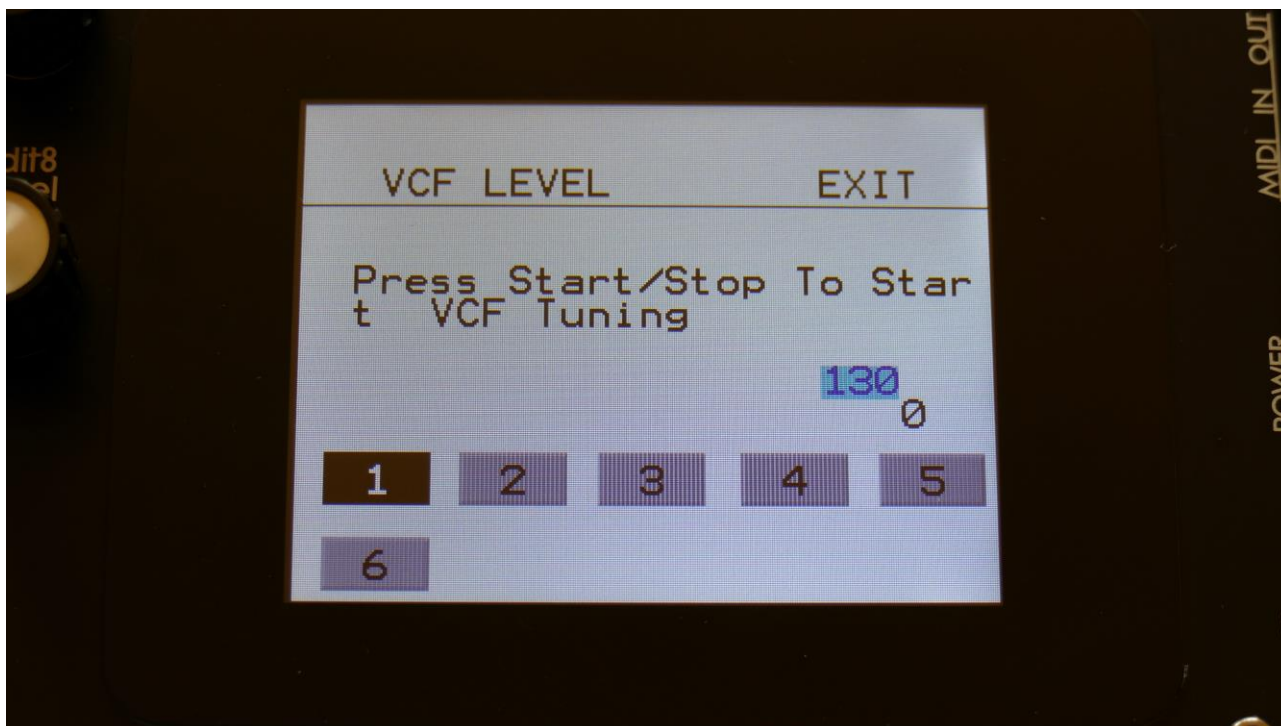
If you feel that the analog filters of your Spazeboard6 has started to sound very different from each other, it might be time to get them tuned. This process is fully automatic, but it does though take around 1 hour, so you should not just do it for fun.

Before you start this process, you should let your LD3 with Spazeboard6 be switched on for at least an hour, to make sure, that the analog circuits is properly heated up.

LD3 with Spazeboard6 does come with its filters tuned from the Gotharman factory, so it should not be necessary to do this for a long time.

You can, at any point of this process, just touch Exit in the upper right corner of the screen, to abort. You will then need to reboot your LD3, in order to make it load the previously stored filter tunings.

From More... page, touch VCF TUNE to enter this page:



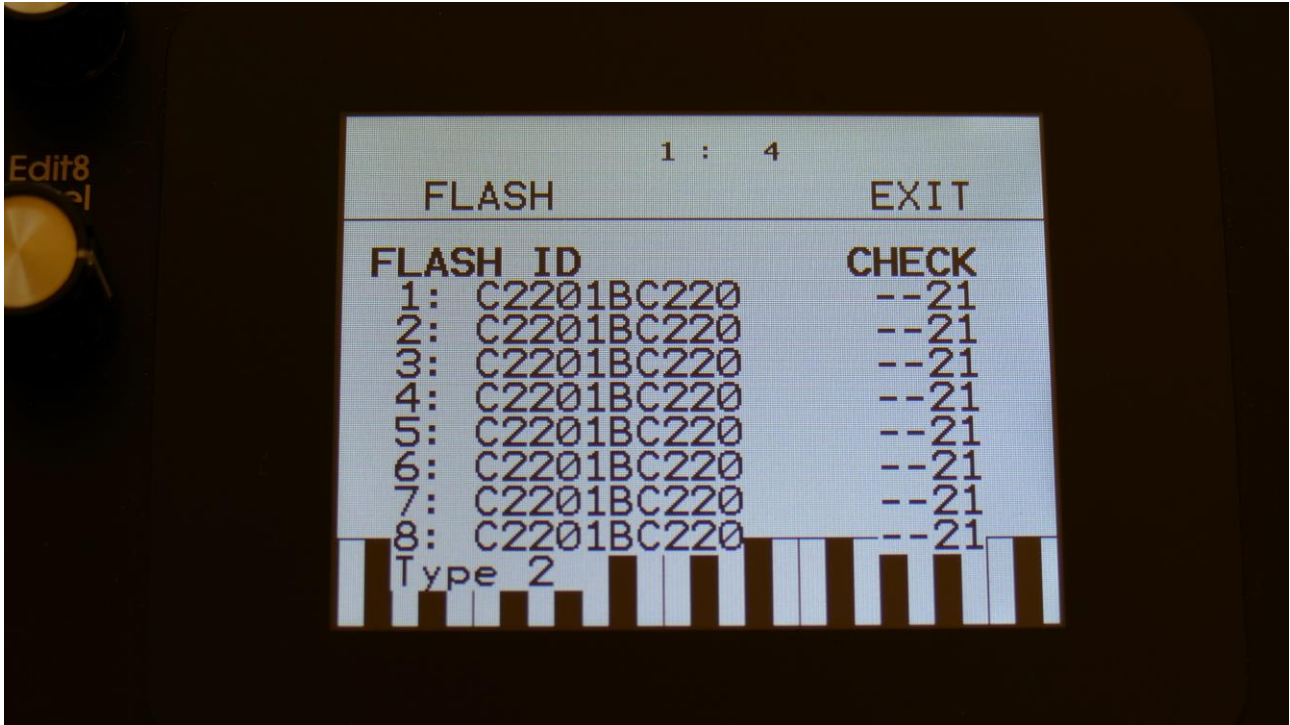
Now push the Start/Stop button, to start the process.

The screen will now update itself from time to time, to show the progress of the filter tuning. Please wait until it jumps back to the MOR page again.

If the filters of your Spazeboard6, after you have tuned them, still not sounds the same, please go to the “TJEK” page, described a little bit later in this section of the manual, and take a picture of the numbers shown there, and send this together with your service request to Gotharman’s.

FLASH Memory Check

From More... page, touch FLASH to enter this page:



On the Xpanded model, this page should look exactly like on the picture. On the un-Xpanded model the numbers will slightly differ, and it will say Type 3 in the bottom.

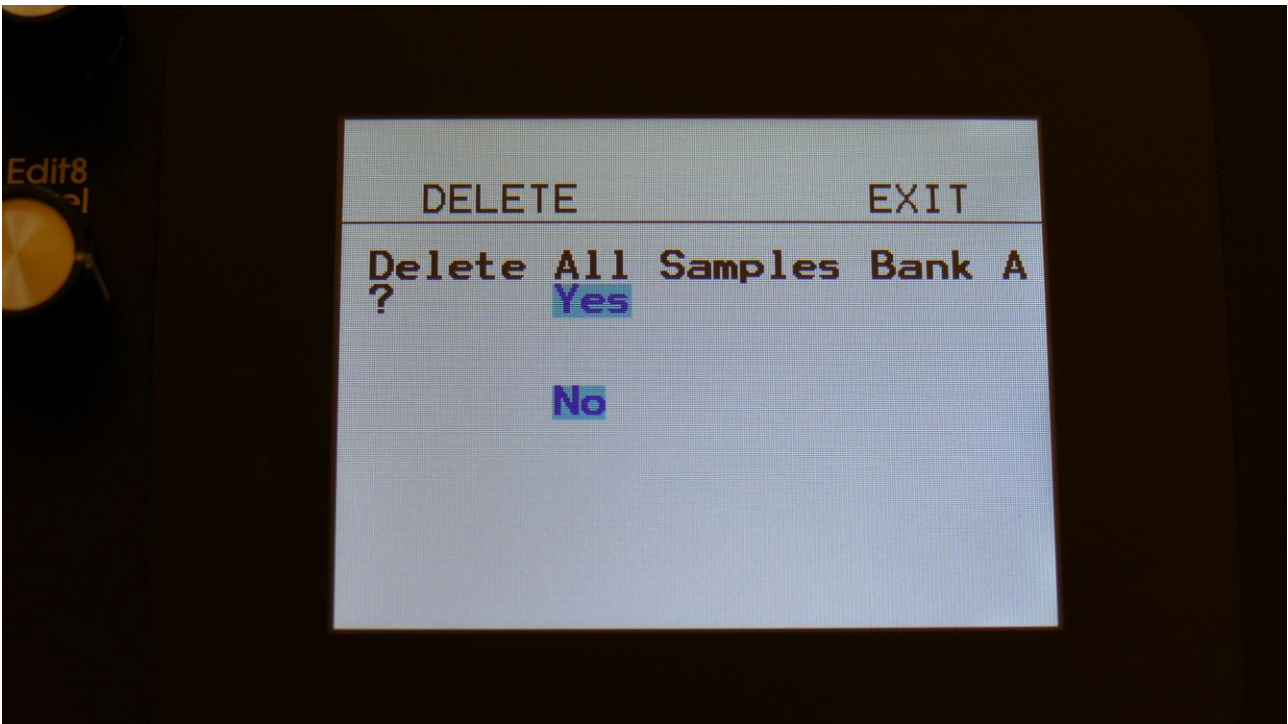
If you should ever experience any issues with missing samples or presets, please take a picture of this screen, and email it to Gotharman's.

Delete a Sample Bank

It is possible to completely erase the sample banks. Please think about it, before you do that, and back up your samplings as described in the USB section later in this manual, since if you don't have a backup, it will not be possible to recover your samplings again.

From More... page, touch DEL SmpA, to erase sample bank A, or touch any of the 3 other choices, to erase the respective sample bank.

LD3 will now ask:



If you are really sure, that you absolutely have to do this, touch "Yes" to start the erase. The sample bank will now be completely wiped out. While it is being erased, LD3's 16 step buttons will run a sequence. When the step buttons shuts off, the erase has been done, and LD3 will return to the More... screen.

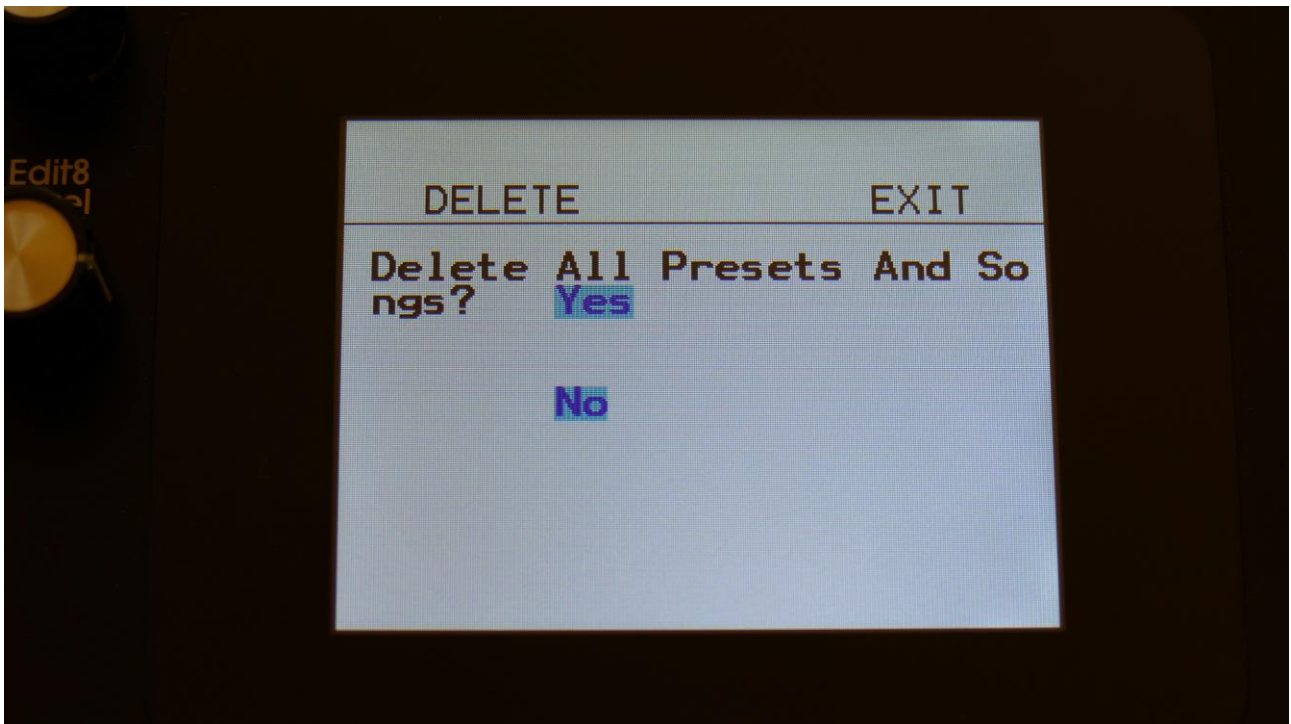
If you don't want to do this anyway, touch "No" to exit.

Delete all Presets and Songs

It is possible to completely erase all presets and songs. To do that, if you are really sure, that you would like to, from More... page, touch DEL Prs.

Before you do this, please back up your presets as described in the USB section later in this manual, since if you don't have a backup, it will not be possible to recover your presets again.

LD3 will now ask:



If you are really sure, that you absolutely have to do this, touch "Yes" to start the erase. All presets and songs will now be completely wiped out. While these are being erased, LD3's 16 step buttons will run a sequence. When the step buttons shuts off, the erase has been done, and LD3 will return to the More... screen.

If you don't want to do this anyway, touch "No" to exit.

C.P.

From More... page, touch C.P. to enter this page:



By turning Edit Knob 1 to 4, you can here make the 2 rectangles produce different colours. This does however not have any practical function at all.

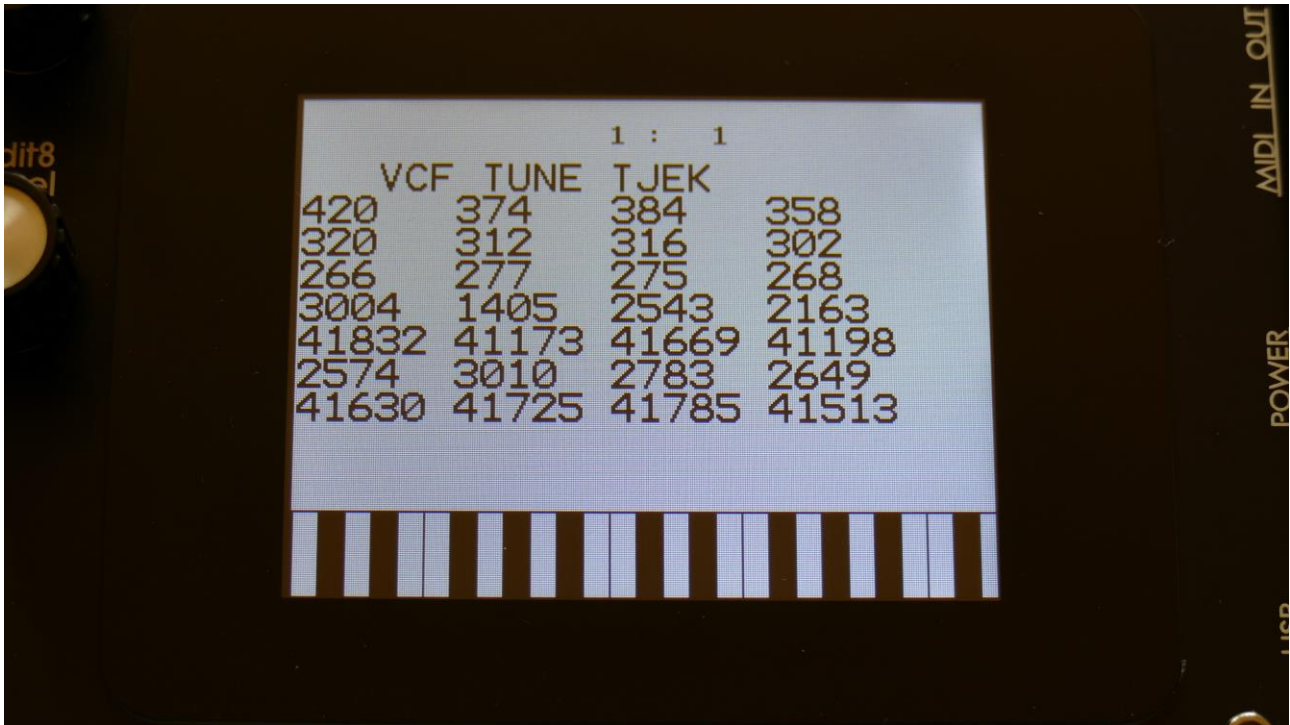
Preset/Song Mode

Touch this to toggle between preset mode and song mode. Song mode is described later in this manual.

TJEK

Entering this page, will make it show a lot of numbers, that Spazeboard6 found in the analog filter tuning process.

From the MOR page touch TJEK, to enter this page:

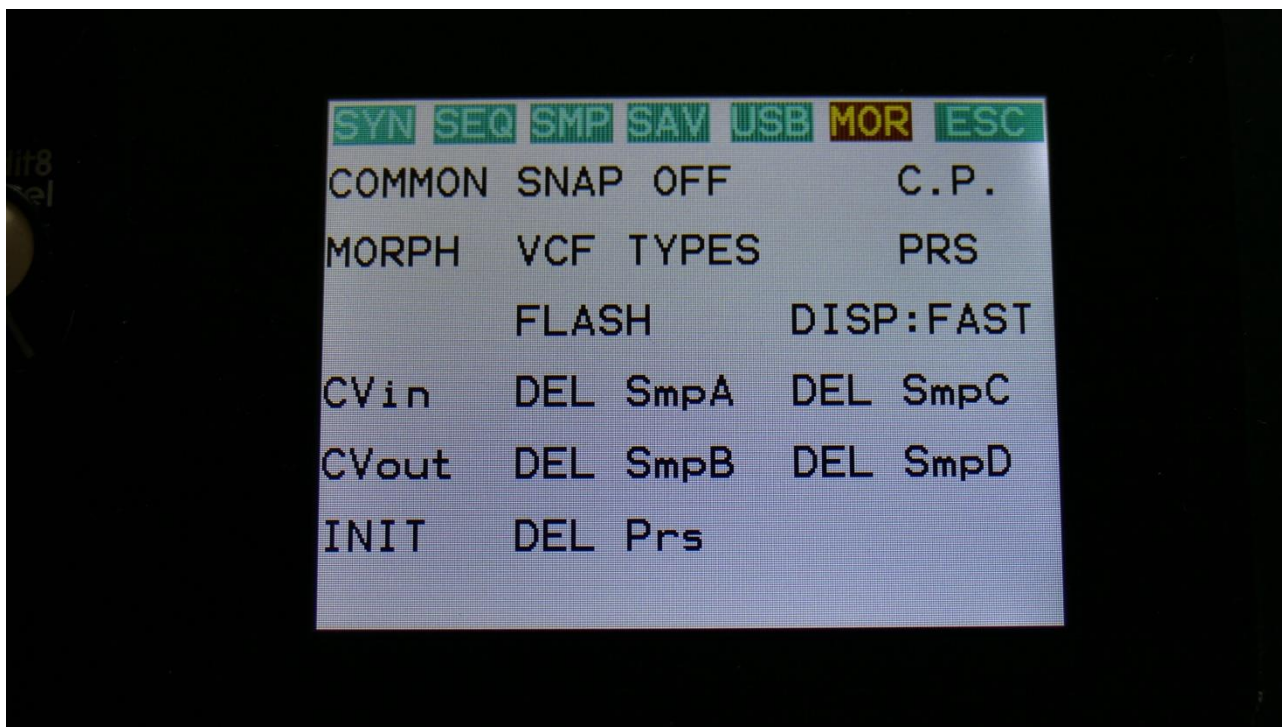


If you think that the analog filter of your Spazeboard6 is malfunctioning, please tune them, as described a little bit earlier in this section, and shoot a picture of these numbers, and send this together with your service request.

Display Update Rate

The display has been set to the maximum possible update rate. I have been testing this as much as possible, and did not discover any display freezes.

If you should discover any display freezes, go to the MOR page:



Touch the text: "DISP:FAST". It will now change to "DISP:SLOW" and go back to the slower display mode, as before.

Please report any display freezes to: info@gotharman.dk .

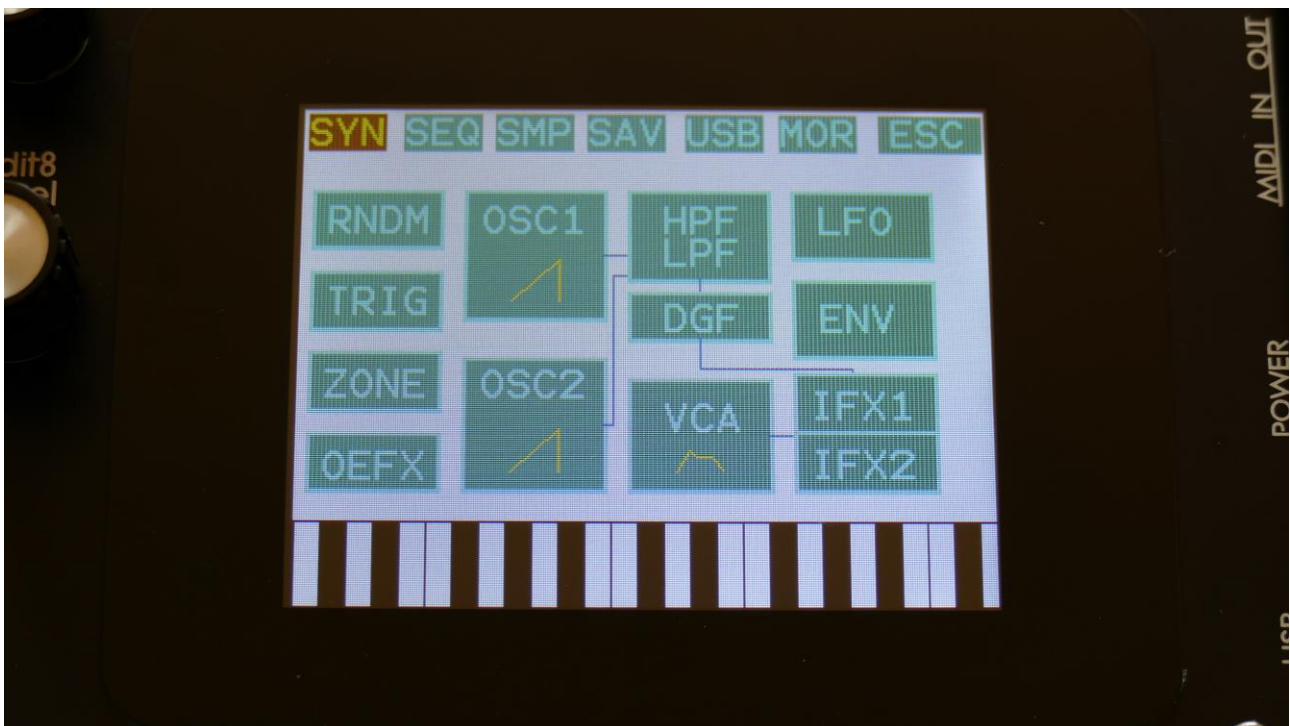
Save Preset

When you have created a new preset, it should be saved for later recall. If you do not save your presets, they will be lost forever, as soon as you select another preset, or turn off your LD3.

A preset that is saved on LD3 contains all the data previously described in this manual:

- All sequencer data
- All synthesizer parts settings
- All effects settings
- All audio bus settings
- All CV in/out settings

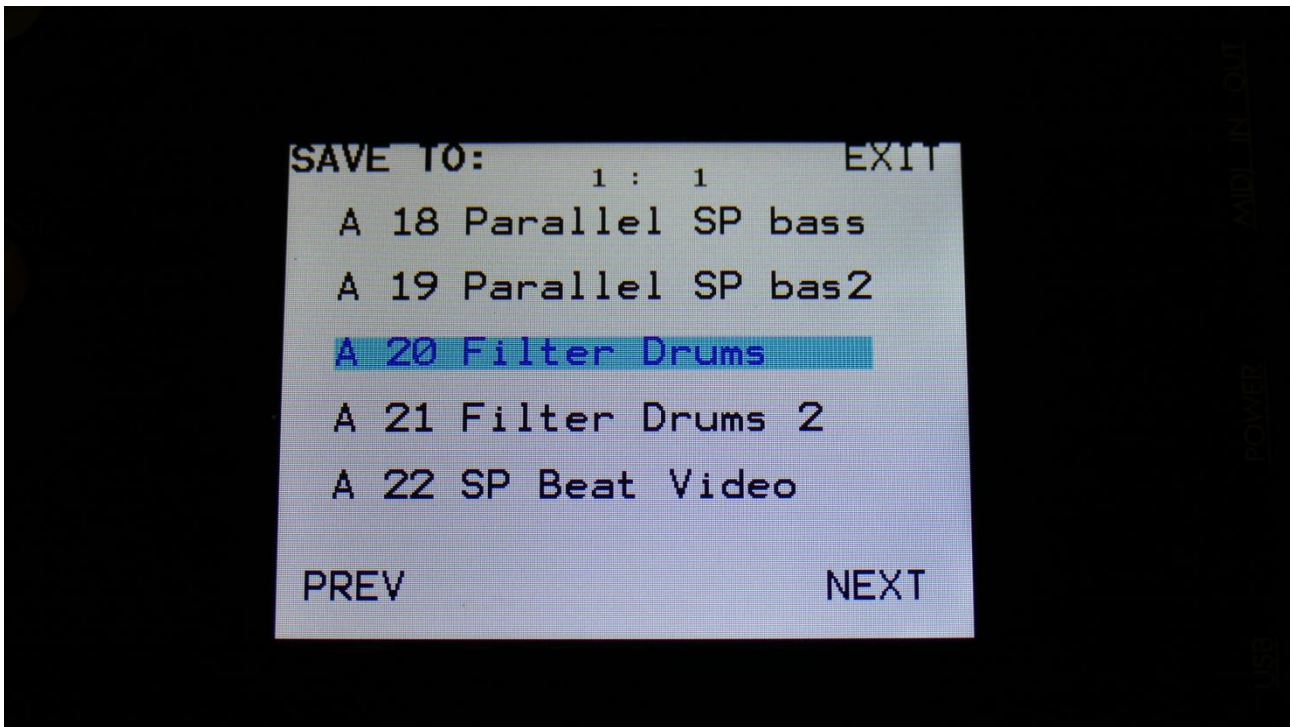
From the Preset Select screen, touch the EDIT field, to enter the Main Synth page:



On the navigation bar at the top of this page, touch SAV.

OR: Hit the Func/mute button, so it lights up, and hit step button 15, marked "Save".

You should now enter this page:

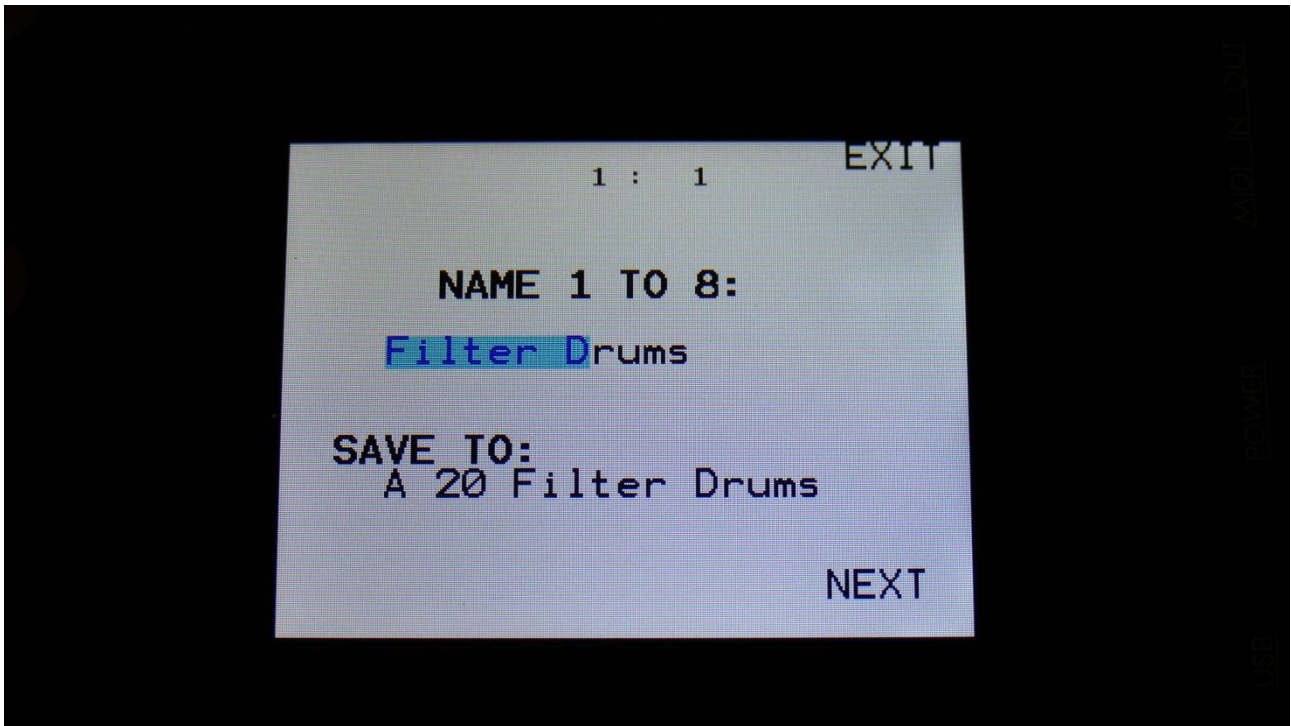


Touch the preset location that you would like to save your new preset to. If it is not shown, touch PREV and NEXT to reach it.

Preset number and name is shown for presets that are already saved. If you save your new preset on such a location, the old preset will be overwritten. On preset locations, where a preset has not yet been saved, the name will be shown as "<empty>".

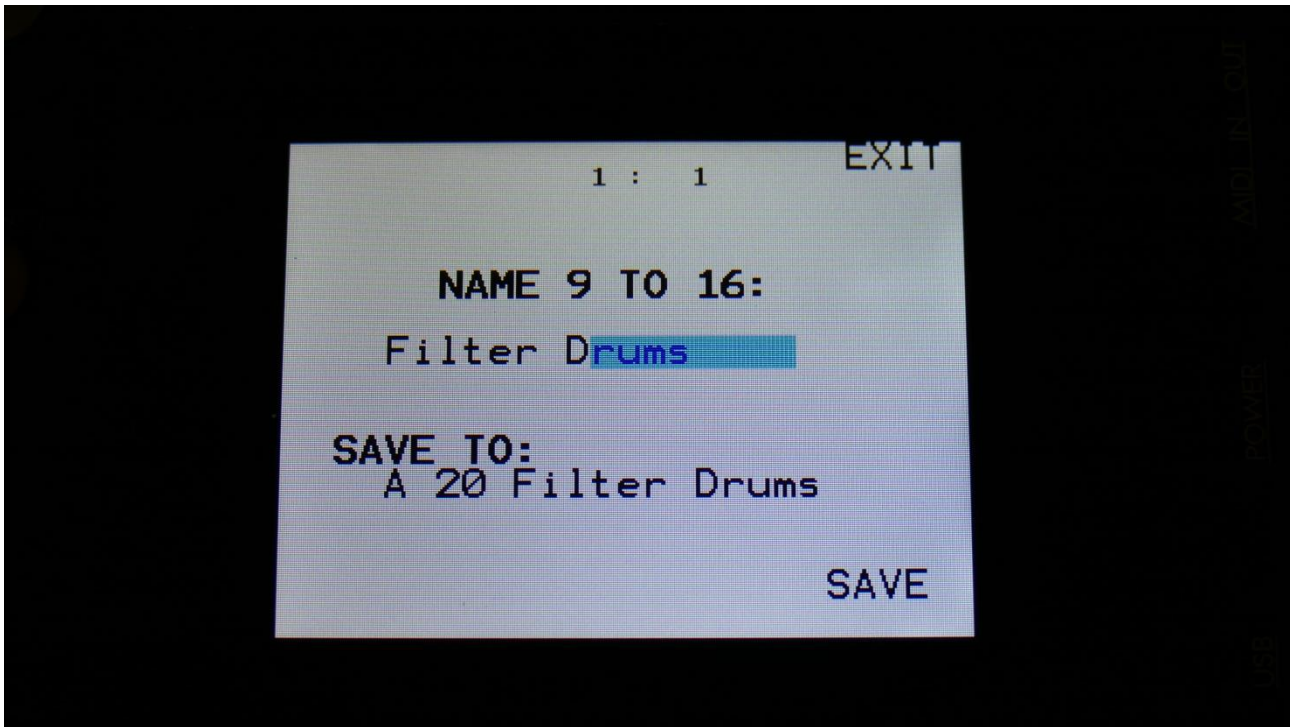
It is also possible to select the destination preset, using the buttons. This is done in exactly the same way, as when selecting a preset, using buttons. Please see the "Operating LD3" section, in the beginning of this manual.

When you touch the preset location, that you would like to save your preset to, you will enter this page:



Turn edit knob 1 to 8 to select the first 8 letters of the name for your new preset.

Touch NEXT when you are done.



Turn edit knob 1 to 8 to select letters 9 to 16 of the name for your new preset.

Touch SAVE when you are done. Your preset will now be saved, and LD3 will exit from the save pages.

Song Mode

It is possible to arrange chained playback of Presets in 1024 Song locations.

Each Song can have up to 128 steps. For each Song step a Preset can be selected, it can be set how many times track one of the preset should play back, until it advances to the next song step, and tracks can be muted/unmuted.

If the last step of the Song is set to "End", playback will stop, when the Song has played back the last preset. If the last step of a Song is set to "Loop", it will jump back to step 1, and continue playback, after the last preset has played.

Songs can be realtime recorded, simply by hitting the "Rec" button on the song main page, then the "Play" button, and then play back the presets that you want in your song.

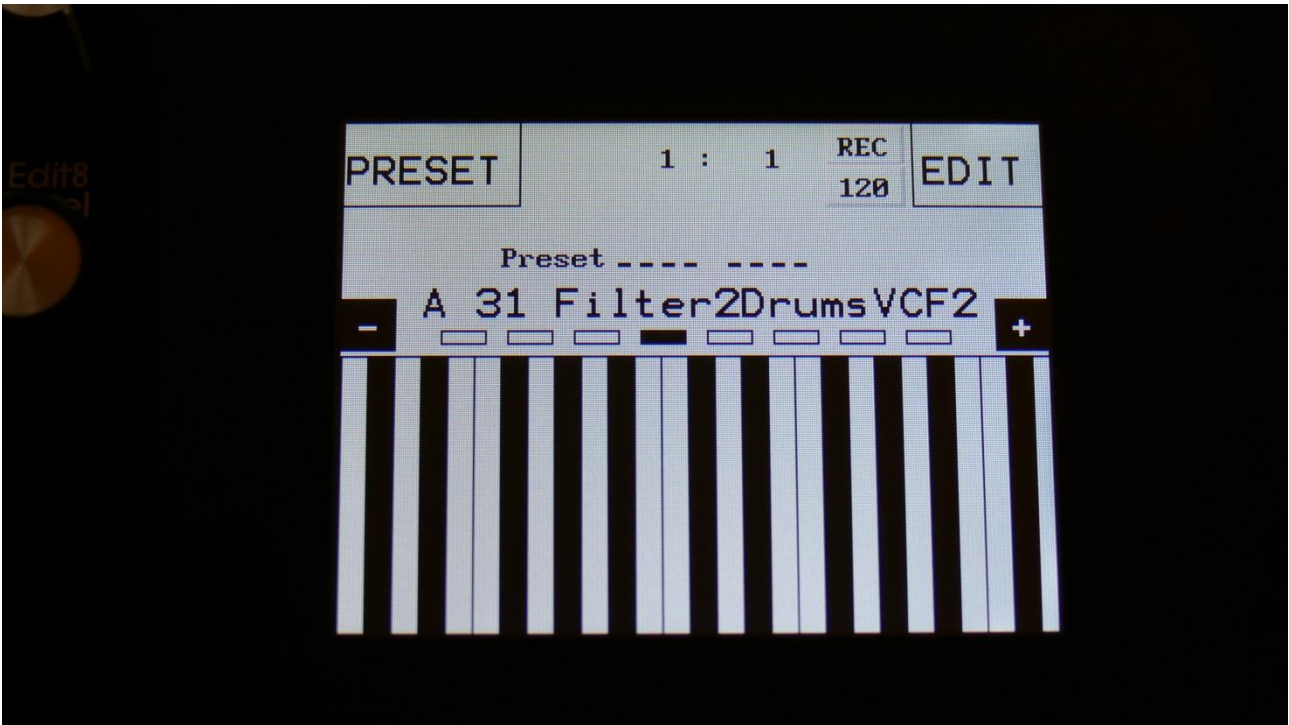
Songs can also be recorded/edited by putting the Preset numbers and number of times to play back, into a list.

Any presets can be used in any songs.

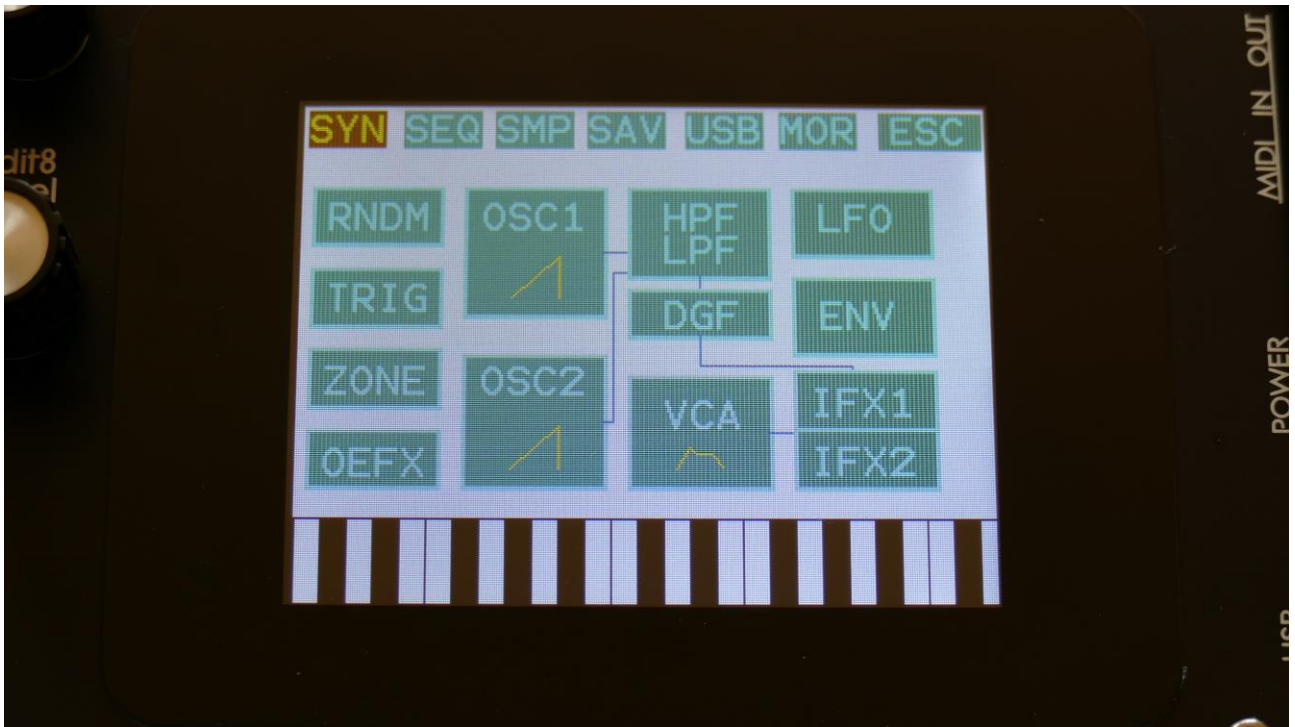
In song mode you still have access to edit all preset synth parameters, in the currently playing preset, but you can't immediately save any changes you make, since in song mode, you can only save the song.

Accessing Song Mode

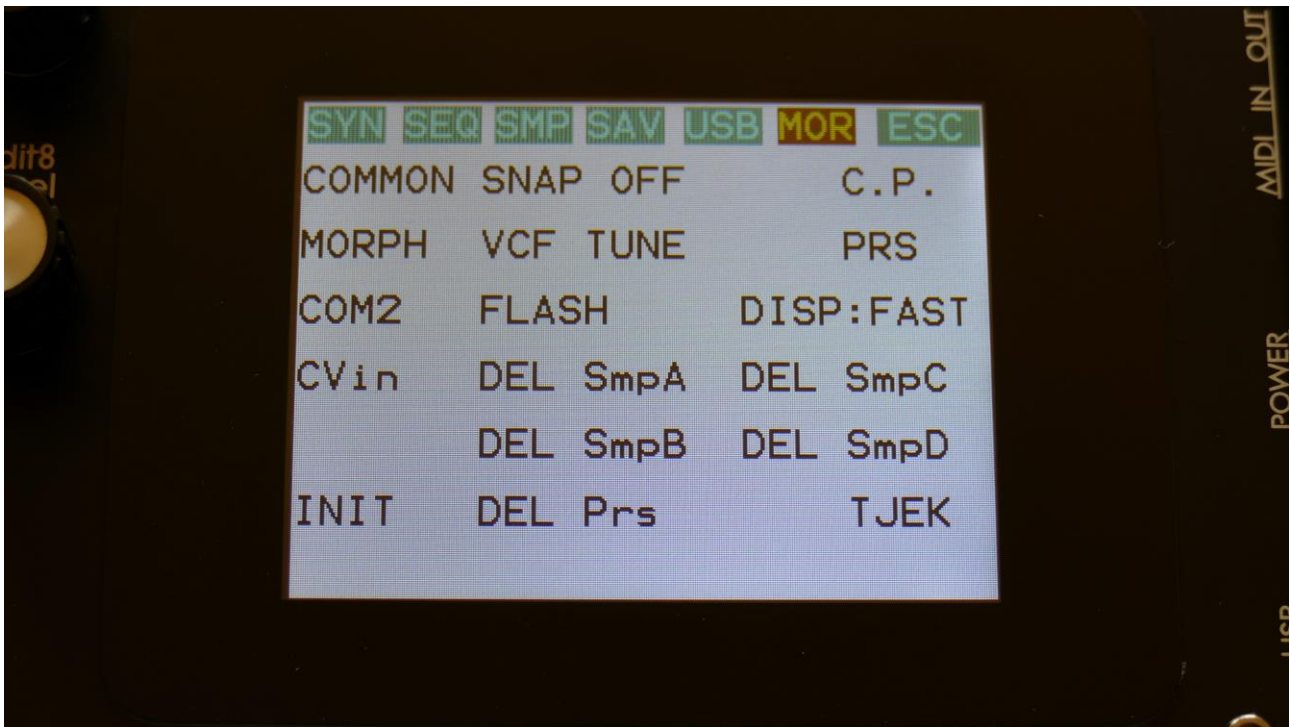
If you would like to make a new song from scratch, it can be a good idea to select the preset you would like to have on the first song step, from the preset select page, before you enters song mode. This is not something you have to do, I have just experienced, when testing this, that it makes things a bit easier. When you select a non-recorded empty song, the last selected preset will automatically be placed on song step 1.



From the Preset Select page, hit the EDIT field.



On the navigation bar at the top of the screen, touch MOR...

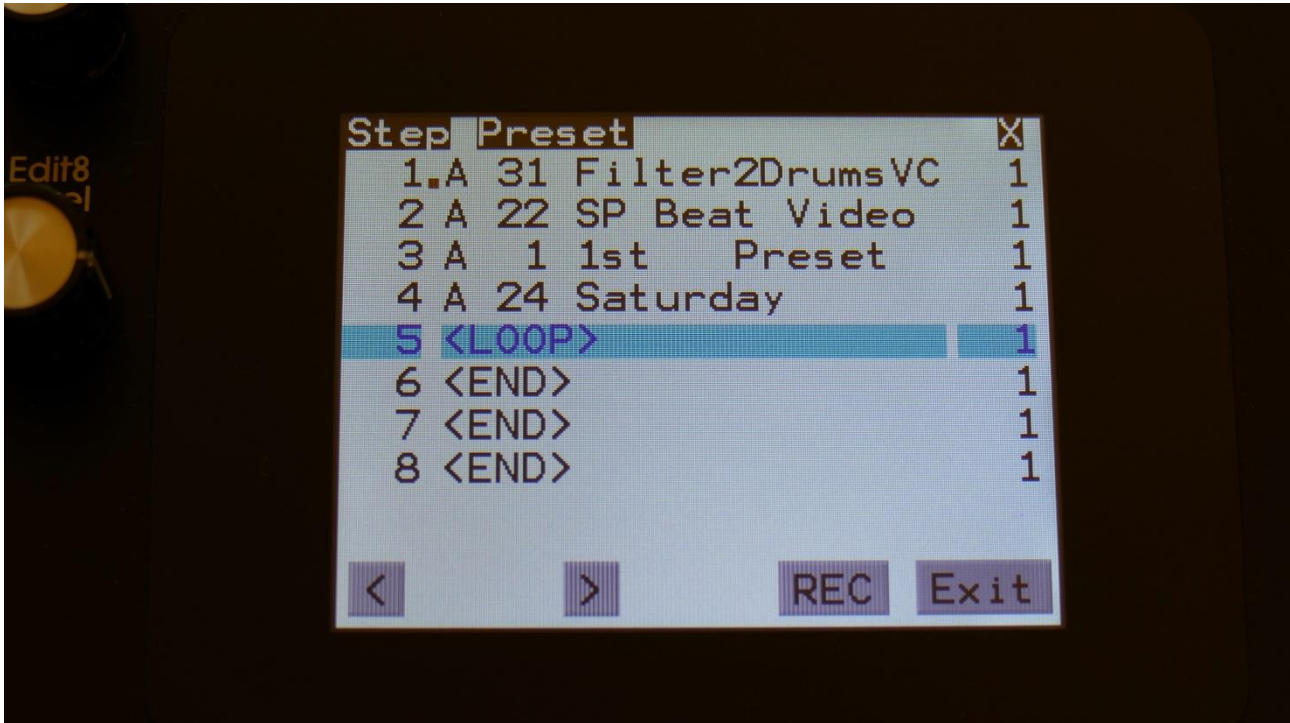


In the far right row, you will find the text PRS. Please touch this, after making sure, that the sequencer is stopped.

Now PRS turns into SONG, and LD3 are in song mode. If you would like to go back to preset mode, just touch SONG.

The Song Edit Page

To enter the Song Edit page, just touch SEQ on the navigation bar at the top of the More... page:



On the Song Edit page, you have an overview of the programmed song steps, and the possibility of editing these.

By touching any of the song steps, you can select this step for editing. The selected step is shown, boxed in blue. The little brown square between the step number and the preset name, is showing what song step is currently being played back.

For each song step, the song step number is shown, the preset number and name, and the number of times track 1 will play back (under the X).

To edit a song step, turn edit knob 1 to select <End>, preset bank A to P, or <Loop>. <End> will make song playback stop, when it reaches that step, <Loop> will make it jump back to song step 1, and start all over again.

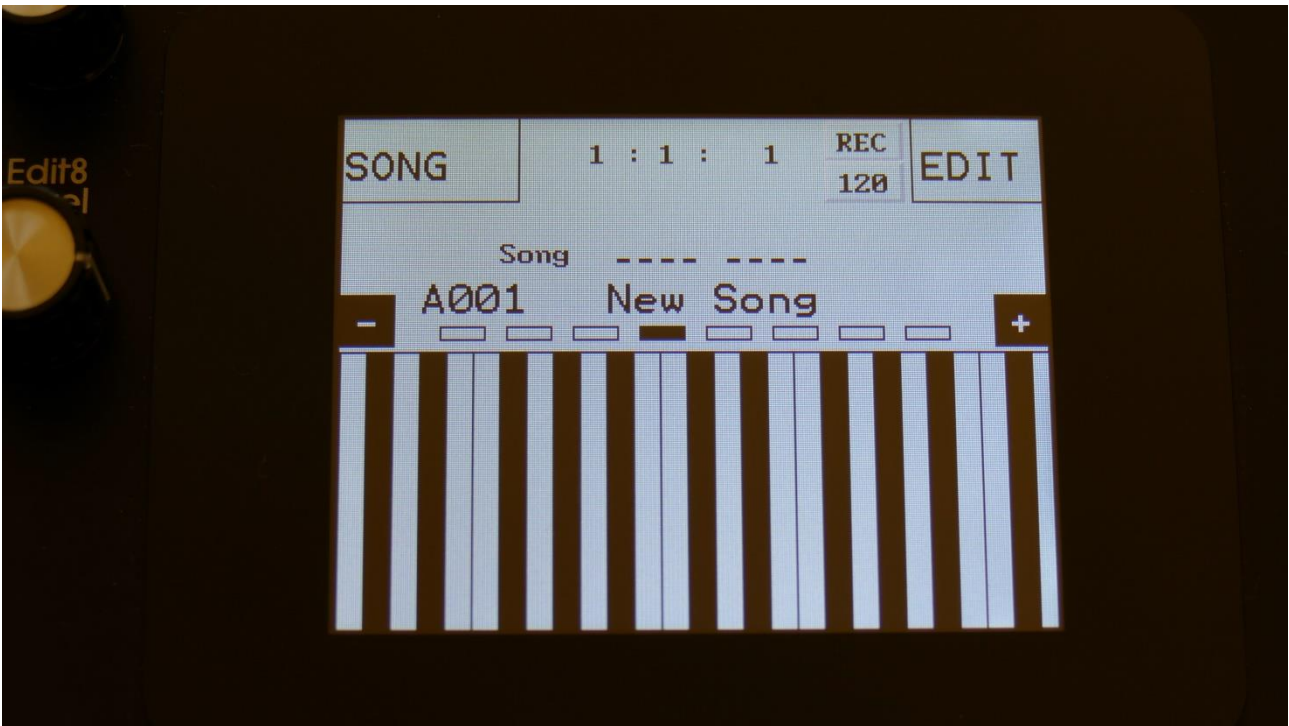
Turn edit knob 2 to select the preset number, and edit knob 3 to select the number of times, you want track 1 to play back, for that step.

Push and hold the Func/Mute button, to mute/unmute tracks on the selected song step, via the step buttons. Only tracks that are unmuted in the preset itself, can be muted by the song sequencer.

8 song steps are shown at a time. Touch the arrows to select the previous or next 8 song steps. The total number of song steps is 128.

Touch EXIT to exit from the Song Edit page, and touch REC to start song realtime recording. This will be explained very soon.

The Song Select Page



When you, in song mode, exit from the Edit Groups pages, you will exit to the Song Select page, instead of the Preset Select page. This reads out "Song" just above the song number and name, instead of the "Preset", to make sure that you know where you are.

On the top of the Song Select page, you might have noticed, that the sequencer position ruler has gained an extra digit. The first digit of this now shows the song step number, that is currently being played back. The 2 other digits is still showing the bar and the beat of the preset, that is currently playing back.

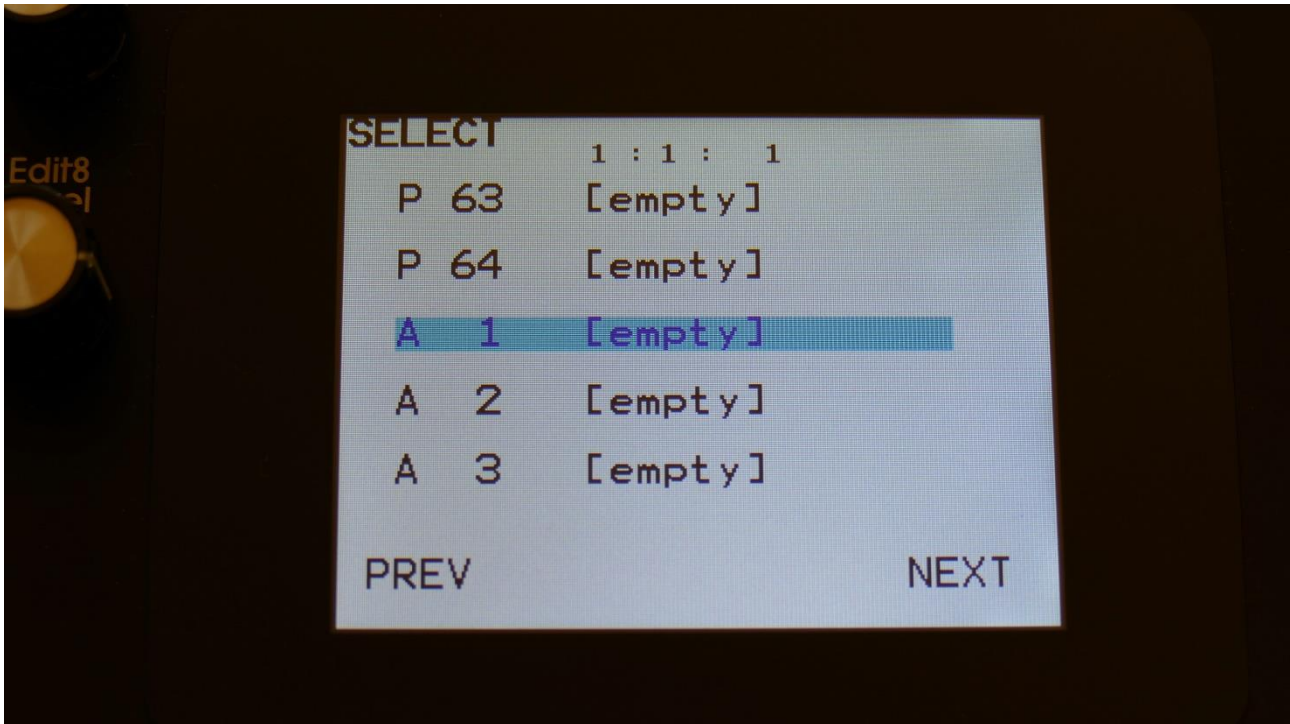
Other changes to this, compared to the Preset Select page, are that the field in the upper left corner, that was used to select preset with, now has been renamed to SONG, and is used to select song with. It is only possible to select another song, while the sequencer is stopped.

Besides from these changes, everything is the same as in preset mode. You can still edit the synth parameters of the currently playing preset.

Selecting a song:

Make sure that the sequencer is stopped.

Touch the "SONG" field. A list of 5 songs near the currently selected song, will now appear:



Touch "PREV" or "NEXT" to view the previous or next 5 songs, and finally touch the song name of the song you would like to select. 1024 songs can be selected, from A01 to P64.

LD3 will now jump back to the main Song Select screen, and show the name of the newly selected song.

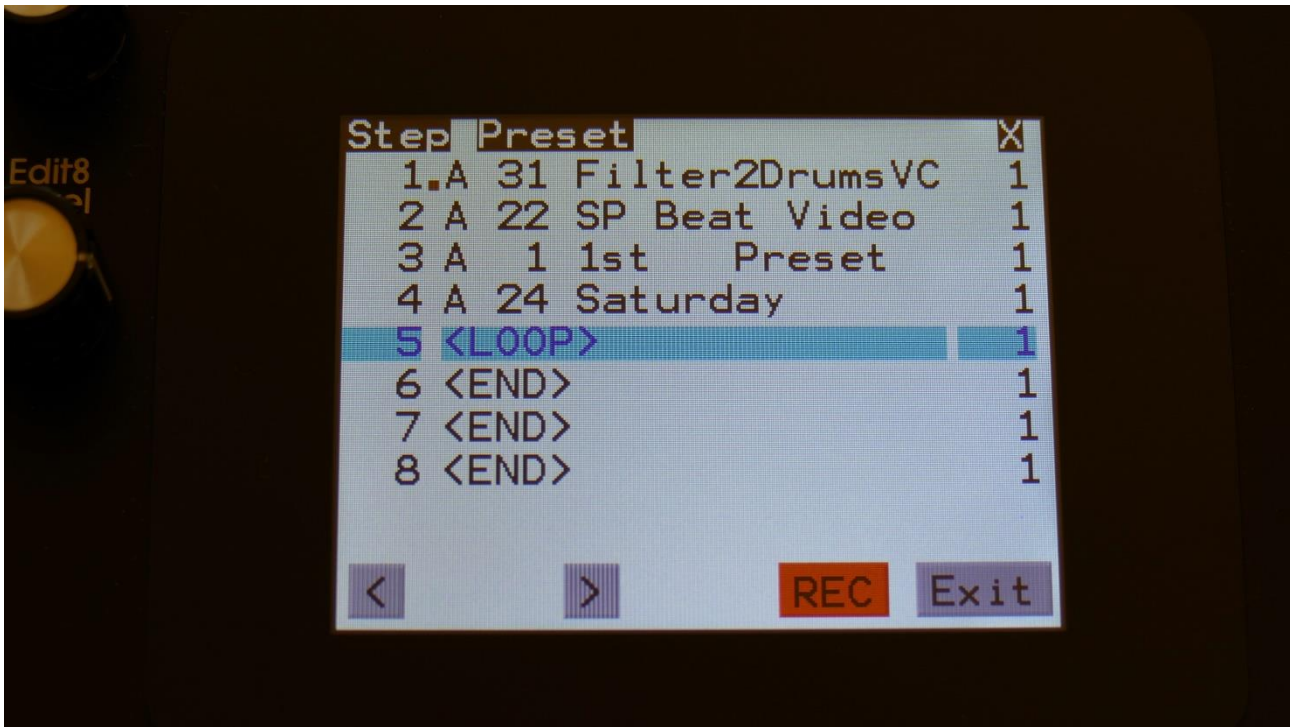
When LD3 is turned off, it will remember which song was selected, and start up with this, when turned on again. It will also remember if it was in preset or song mode, and start up in the same mode.

Song Realtime Recording



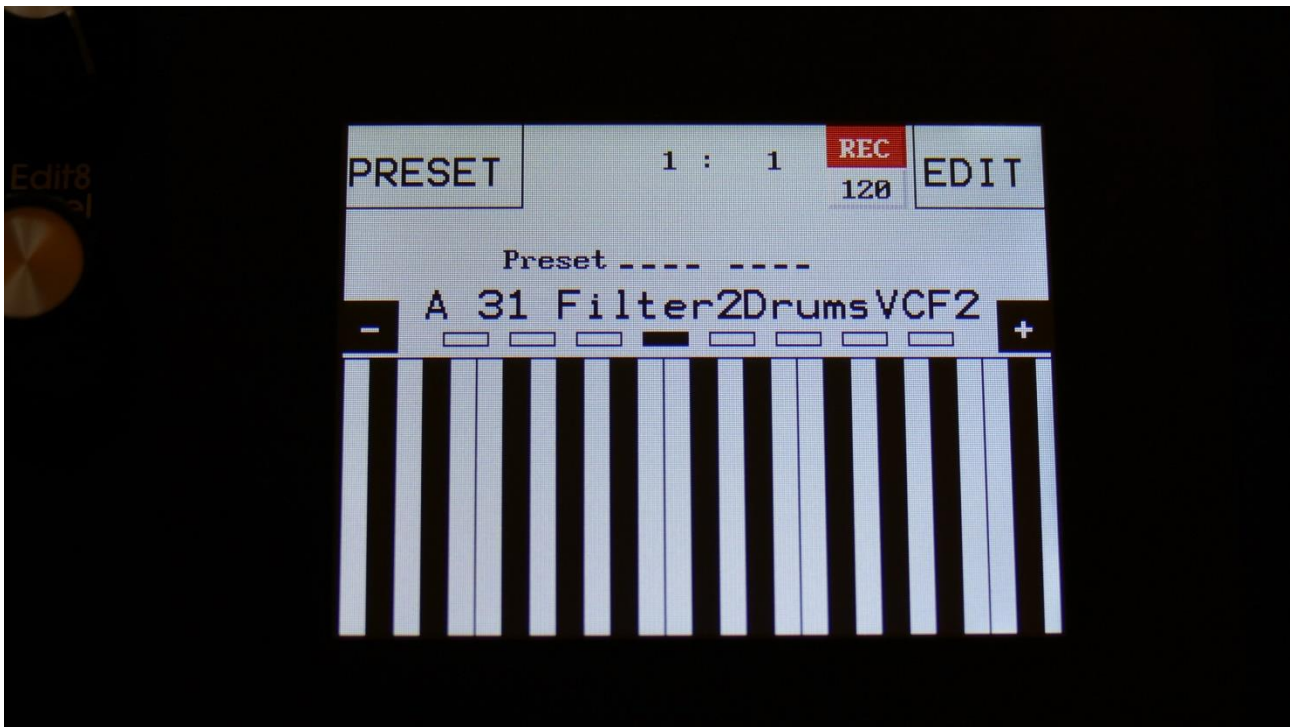
On the Song Select page, select the song you would like to record. 1024 songs are available, from A00 to P64.

Now go to the song edit page.



Make sure that the sequencer is stopped, and push the “**REC**” button, so it turns red.

Now exit to the Preset/Song select screen.



LD3 will now show the Preset select page. The REC indicator on the preset select page, has now turned red, indicating that LD3 is in recording mode. Song realtime recording can also be switched on or off, by touching the REC indicator.

If you selected an empty song, the last preset you selected, before you jumped to song mode, will be shown. If you selected a song that was already recorded, the preset on the selected step of this song will be shown.

Now, if the shown preset are the preset you would like to have on step 1 of your song, you can proceed recording your song, by pressing the Start/Stop button.

If you would like another preset on step 1 of your song, please select this, using the preset select field, and then push the Start/Stop button, to start the recording of the song.

The preset you selected will now start to play back.

From here you can now select other presets, at the time you want these to play back in your song.

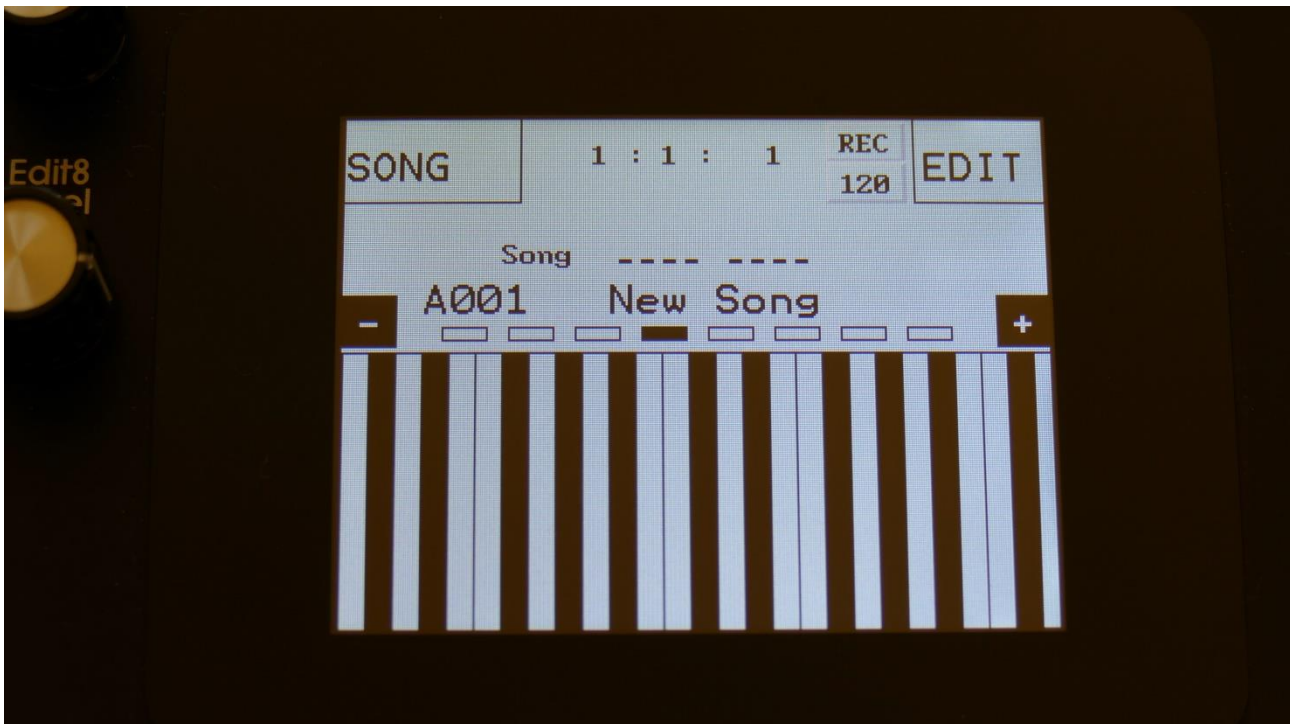
Preset changes will only be registered and recorded, when track 1 of the currently playing preset reaches its end and starts over. When the changes has been registered, the song step number in the position ruler will increment by one.

While any preset is playing back, it is possible to mute/unmute the sequence tracks. Push and hold the Func/Mute button, and switch tracks on or off, using the step buttons.

To advance to the next song step, to have other tracks of the same preset muted/unmuted, simply reload the preset, by pushing and holding the Morph Set button, while pressing the Start/Stop button.

When you are done changing presets, and you think that your song is ready, hit the Start/Stop button to stop song recording.

Go back to the song edit page, and touch the REC button, so it turns grey again, to exit from song realtime recording, or simply exit to the preset/song select page, and touch the REC indicator.



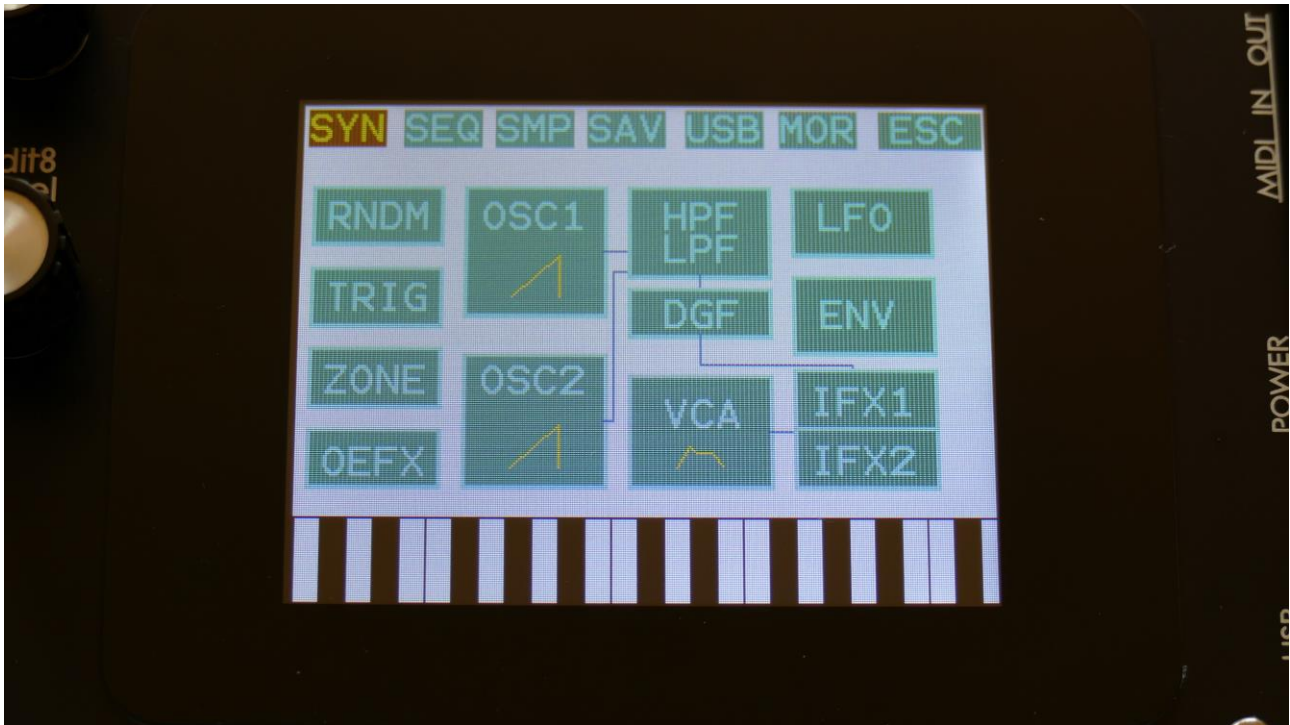
To listen to the song you have just created, just hit the Start/Stop button.

If there are anything you would like to edit, go to the Song Edit page, as described earlier.

If you would like to keep your work, you should save your new song.

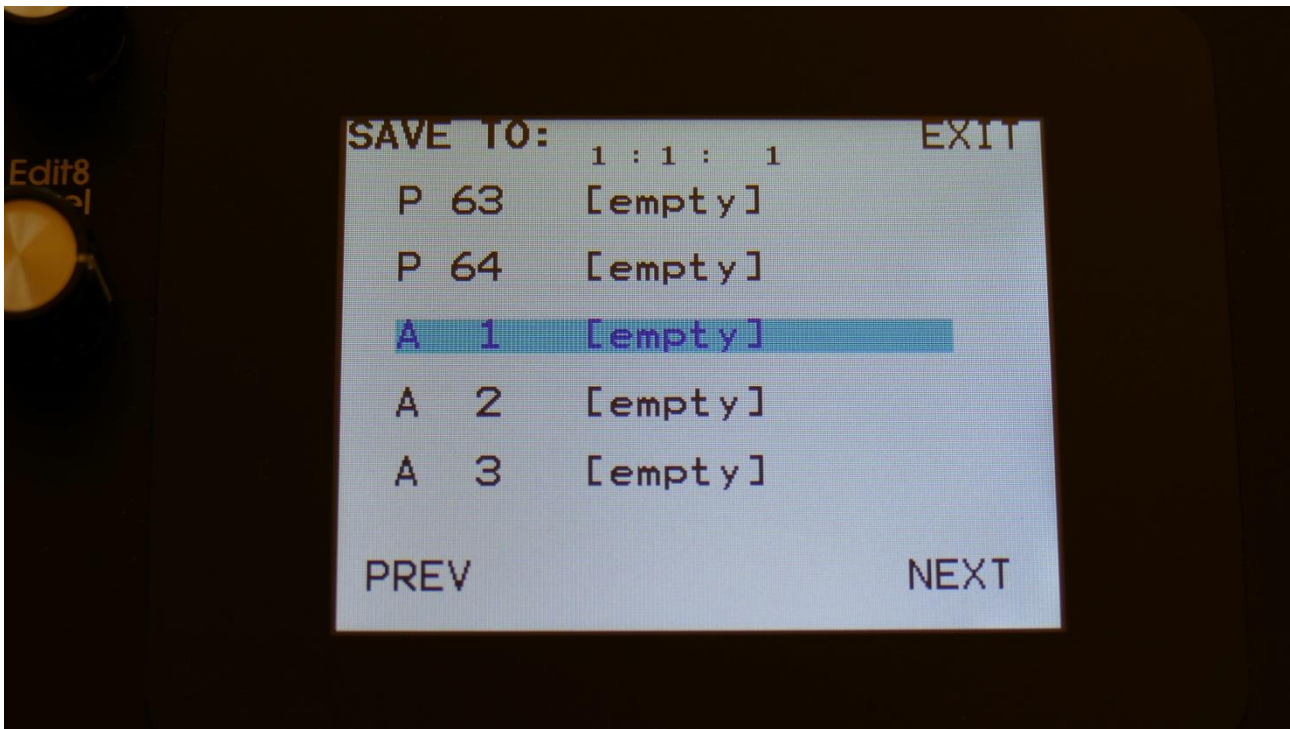
Save Song

LD3 songs are not saved within the presets, so they must be saved separately. To do so, when in song mode, from the song select page, touch the EDIT field to enter the Synth parts page:



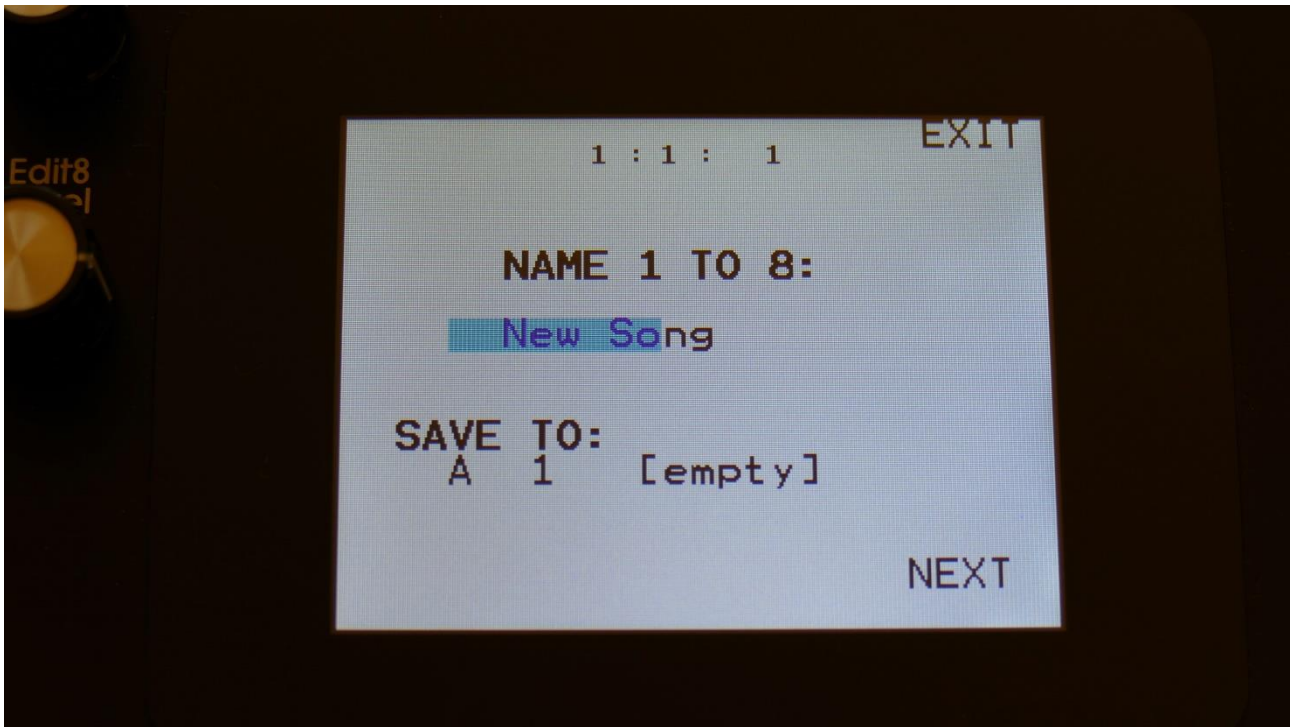
Touch SAV.

You should now enter this page:



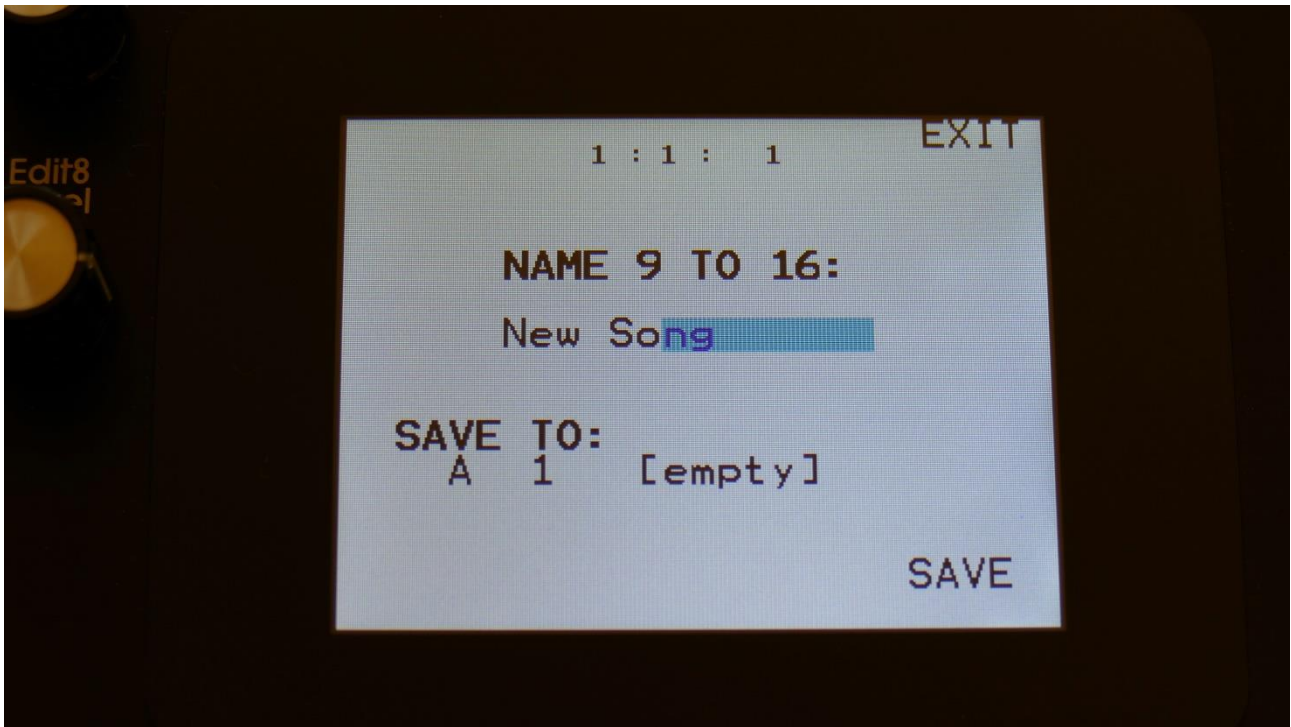
Touch the song location that you would like to save your new song to. If it is not shown, touch PREV and NEXT to reach it.

Song number and name is shown for songs that are already saved. If you save your new song on such a location, the old song will be overwritten. On song locations, where a song has not yet been saved, the name will be shown as "<empty>".



Turn edit knob 1 to 8 to select the first 8 letters of the name for your new song.

Touch NEXT when you are done.



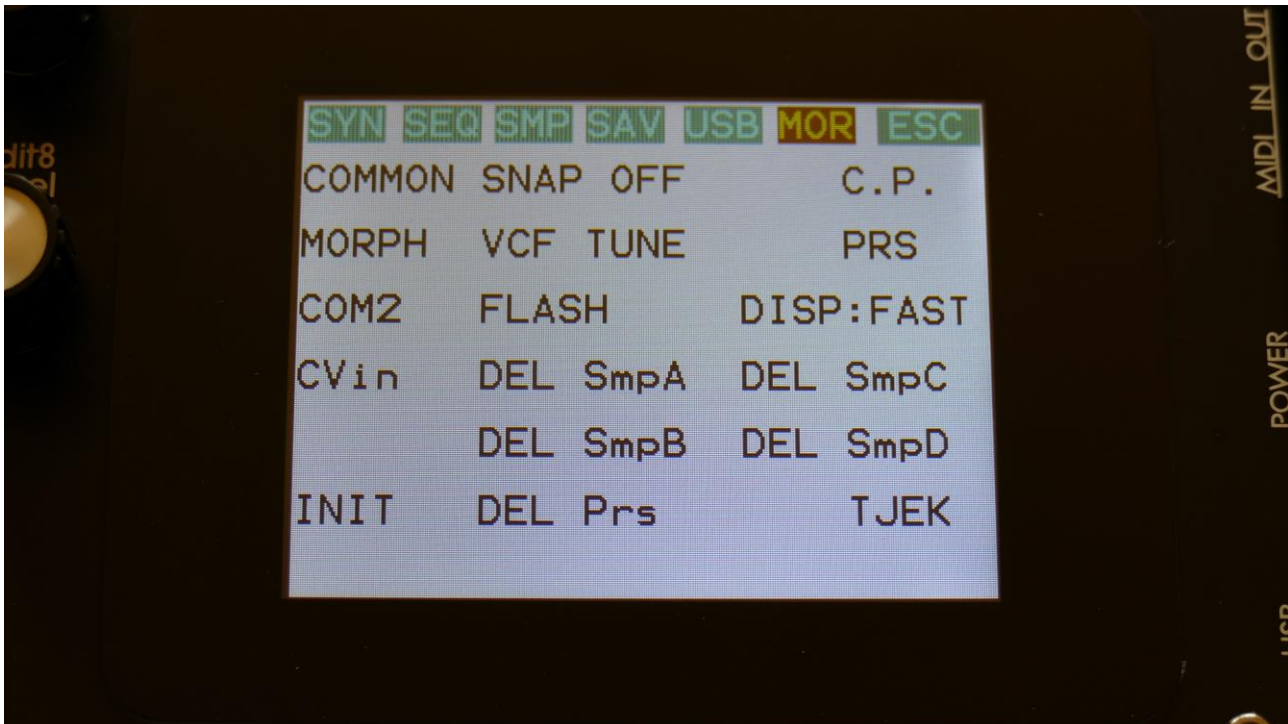
Turn edit knob 1 to 8 to select letters 9 to 16 of the name for your new song.

Touch SAVE when you are done. Your song will now be saved, and LD3 will exit from the save pages.

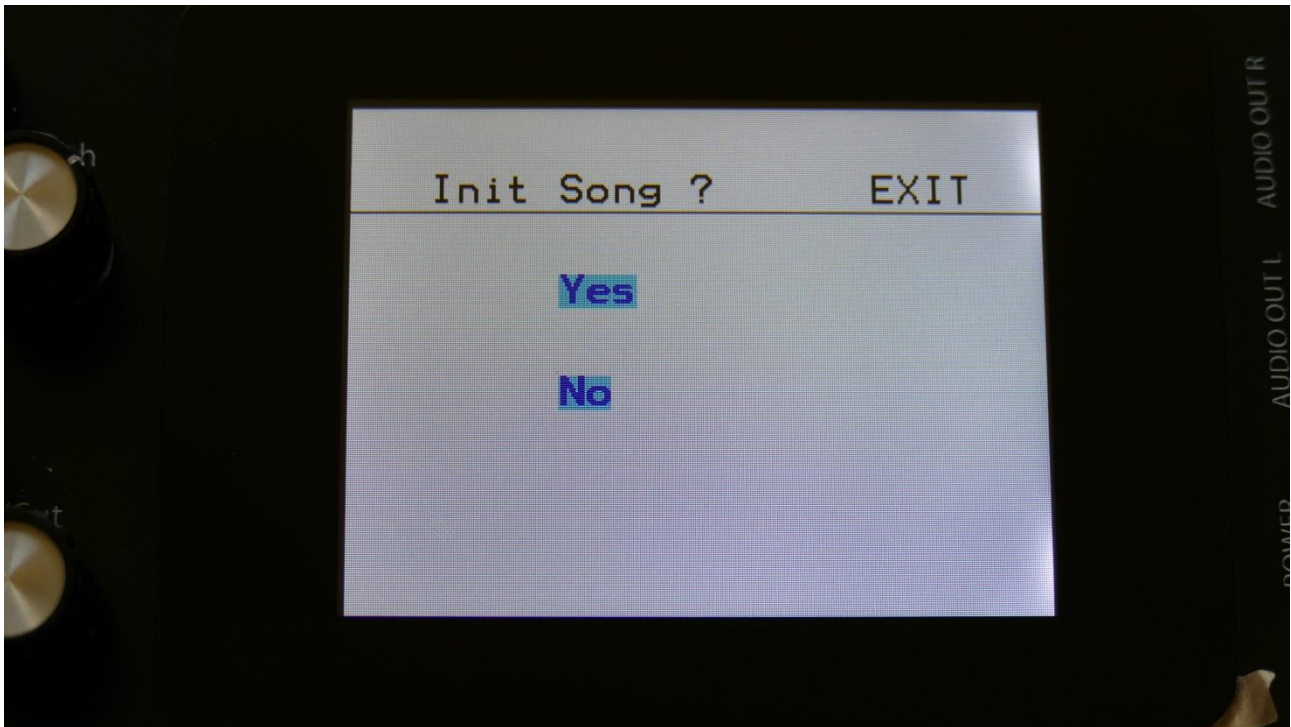
Initializing a Song

If you would just like to start all over with an empty song, it is possible to initialize it.

To do so, in song mode, access the MOR page:



From here, hit INIT.



LD3 will now ask you to confirm.

Press "Yes" to initialize the currently selected song and return to the previous page, or press "No" just to exit, without initializing.

Please notice, that when initializing a song, this is only done in LD3's temporary song RAM, NOT on the FLASH memory, so if you did this by mistake, or you regretted doing it, you can always recall the song, by selecting another song, and select this again. Then nothing will be lost. It is not until you save the song, that permanent changes is done.

Copy/Paste

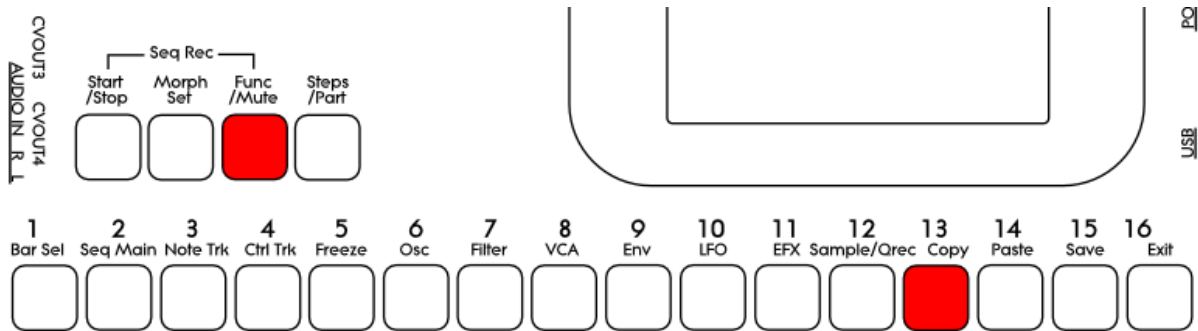
On LD3 it is possible to copy morph layer A, B or the parameter setting at the position of the Morph knob. It is possible to copy one or two layers of the whole preset, or only a section of the preset, like a synth part, the settings of a filter, a sequencer track, and much more.

Whatever is held in the copy buffer can then be pasted to another preset or another section.

If you, for instance, copy a whole preset, or a whole part, and then goes to a specific filter page to just paste the filter settings, only the settings in the copy buffer from the same part number will be pasted. If you would like to copy, for instance, the filter settings from part 3 in one preset, to part 7 in another preset, you will have to go to the specific part filter page, to copy, and then go to the destination filter page to paste.

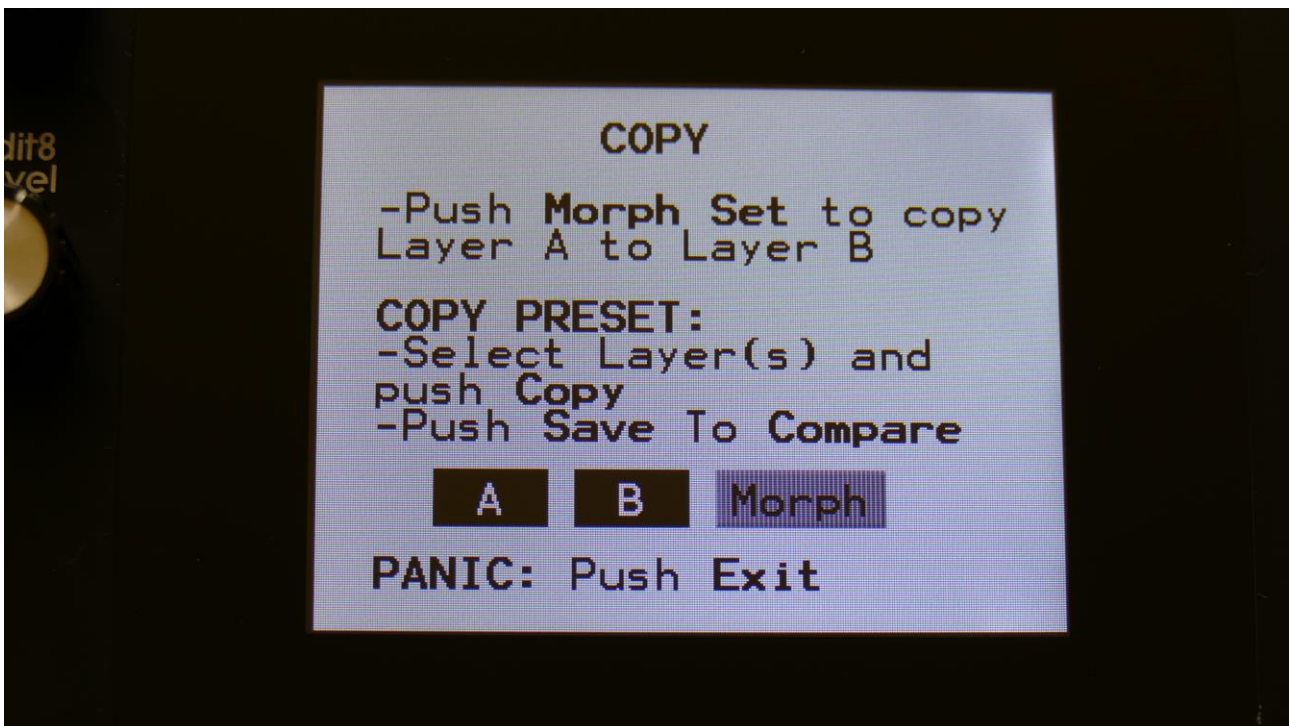
To copy anything, first go to the page, that will make LD3 copy the desired settings. Possibilities are:

- Preset select** page: LD3 will copy one or two layers of the whole preset.
- Synth part** page: LD3 will copy one or two layers of the selected part. Included in this are the oscillator, the analog and digital filters, VCA and Envelope settings.
- Oscillator** page: LD3 will copy one or two layers of the currently selected part, but only paste the copied oscillator settings.
- VCF** page: LD3 will copy one or two layers of the currently selected part, but only paste the copied analog filter settings.
- Digital Filter** page: LD3 will copy one or two layers of the currently selected part, but only paste the copied digital filter settings.
- VCA** page: LD3 will copy one or two layers of the currently selected part, but only paste the copied VCA settings.
- Insert Effect** page: LD3 will copy and paste one or two layers of the currently selected Insert Effect.
- Envelope** page: LD3 will copy one or two layers of the currently selected part, but only paste the copied envelope settings.
- LFO** page: LD3 will copy and paste one or two layers of the LFO settings, of the currently selected LFO.
- Output EFX** page: LD3 will copy one or two layers of the EFX settings, of the currently selected EFX.
- RNDM, TRIG and ZONE** pages: Nothing will be copied.
- Sequencer overview** page: One or two layers of all the sequencer tracks will be copied.
- Sequencer track** page: One or two layers of the currently selected sequencer track will be copied.



When you are located at the right page, push and release the Func/Mute button, so that it lights up, to enter the function buttons. Now hit step button 13 (Copy).

The display will now show some instructions:



It will tell you how to copy layer A to layer B, and right below this, it will tell you, what it will copy. In the example, the copy function was entered from the preset select page, and LD3 tells you, that it will copy the whole preset.

If you had entered the copy page from a filter page, it would have said "COPY SYNTH PART 1" instead.

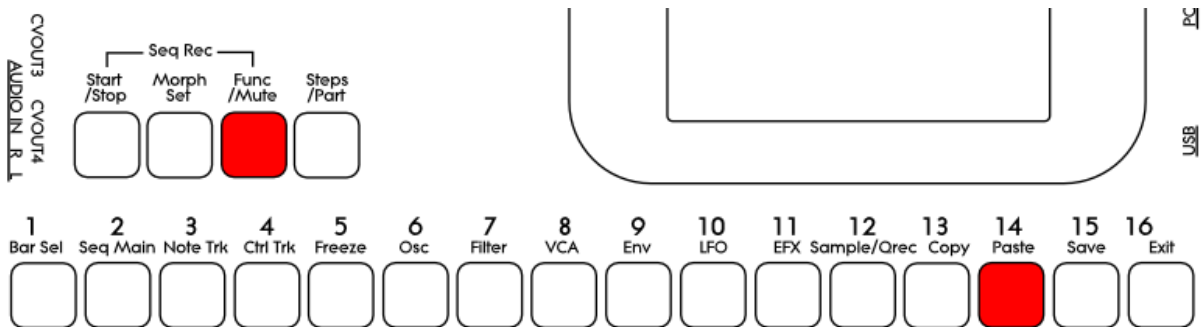
Now select/deselect the desired layers, by touching the "A", "B" and "Morph" touch buttons. If a layer button is black, it means that this layer will be copied, if it is grey, the layer will not be copied. It is only possible to copy either layer B or the morphed layer. It is not possible to copy

both of these at the same time. Layer A can be switched on or off independently of the B/Morph selection. If 2 layers are copied, LD3 will also paste to 2 layers. If only one layer is copied, LD3 will paste to the layer selected by the Morph Set button.

To copy, hit step button 13 (Copy) again.

Now, when the function buttons are active (the Func/Mute button lights up), every time you enter a page, where the copy buffer holds something, that can be pasted, step button 14 (Paste) will light up.

To paste something, navigate to the desired page, where you would like to paste the settings. Make sure that step button 14 (Paste) lights up. Hit step button 14 (Paste). Your settings has now been pasted.



Sample Record and Edit

To get samples into LD3, for using them in the synth section samplers, you can either import .wav files from a USB drive, or record your own samples from LD3's audio inputs or outputs. This section will describe the latter. For how to import .wav files, please see the USB section of this manual.

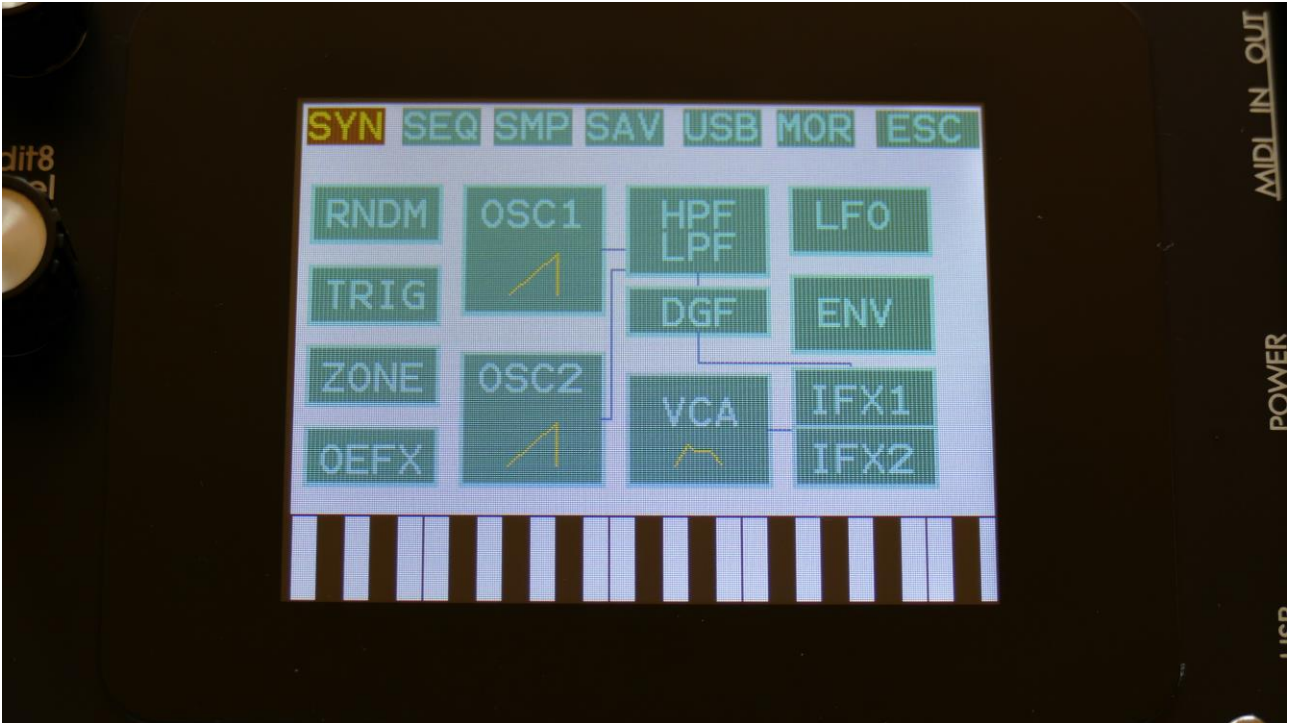
Samples are recorded and played back in mono or stereo.

LD3 has 4 banks of FLASH memory for storing samples, Bank A, B, C and D. Bank A holds up to 18 minutes of mono samplings (43 minutes in the Xpanded model), maximum 2048 samplings, Bank B, C and D each holds up to 25 minutes of mono samplings (50 minutes in the Xpanded model), maximum 2048 samplings – A total of 8192 samplings/93 minutes or 193 minutes. One sampling can maximum be 50 minutes long. Sample times in stereo are half of the mono sample times.

All samples that are imported or recorded, stays in LD3's FLASH memory, even after a power off. Since they are played back directly from the FLASH memory, there are no loading times. All LD3's samplings are immediately available, right after power on, and can be selected in the synth oscillator section, just like synth waveforms.

Recording a sampling

From the Preset Select screen, touch the EDIT field, to enter the Main Synth page:

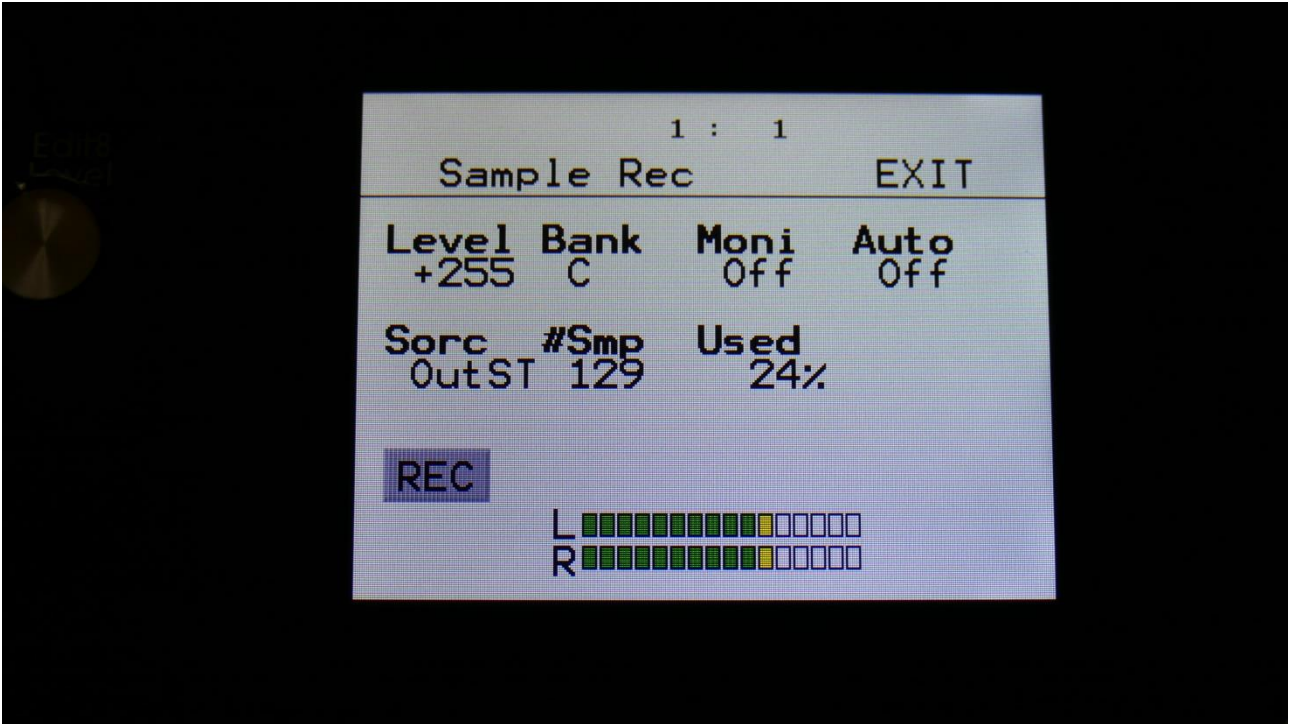


Touch SMP at the top navigation bar.



You will now enter the Sample Edit page.

Touch the REC touch button, to enter the Sample Rec page.



OR, from any page:

-Push and release the Func/Mute button, so it is lighting up.

-Push step button 12, marked "Sample/Qrec".

This will also get you to this page, just much faster.

Since LD3 with Spazeboard6 can only sample the stereo outputs, the Moni parameter is not present on this.

Select the record source, by setting the Src parameter.

Choices are:

OutpL : Left audio output mono

OutpR : Right audio output mono

OutST : Left and right audio outputs stereo

Now select the sample bank, that you would like to record to, setting the Bank parameter to bank A, B, C or D.

When choosing a bank, the **#Smp** parameter will show how many sampling that the selected bank already holds, and the **Used** parameter will show how many percent of the sample bank, that are already in use.

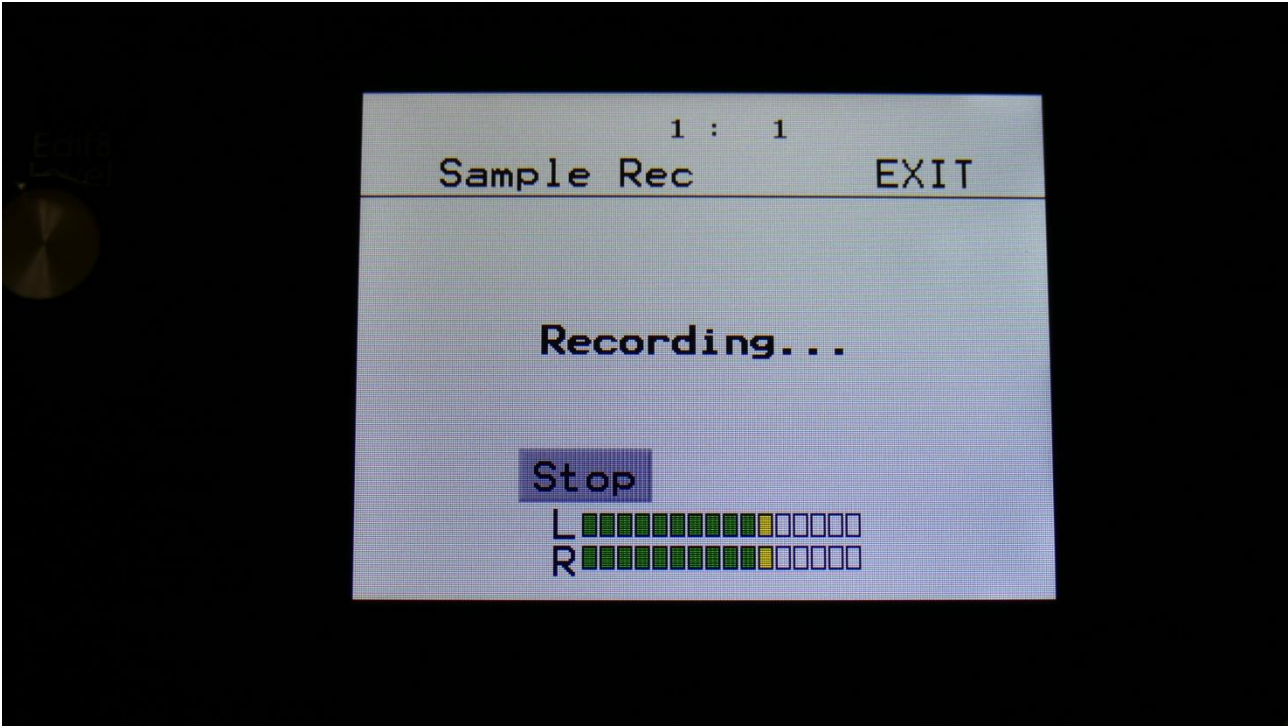
For optimal results, set the level parameter to +0, and adjust the level of the record source, so that the 2 VU-meters in the bottom of the screen does not get red.

You should only adjust the level parameter, to other values than +0, if it is not possible to adjust the level directly on the source.

When you are ready to record the sampling, touch the "REC" button in the lower left corner.

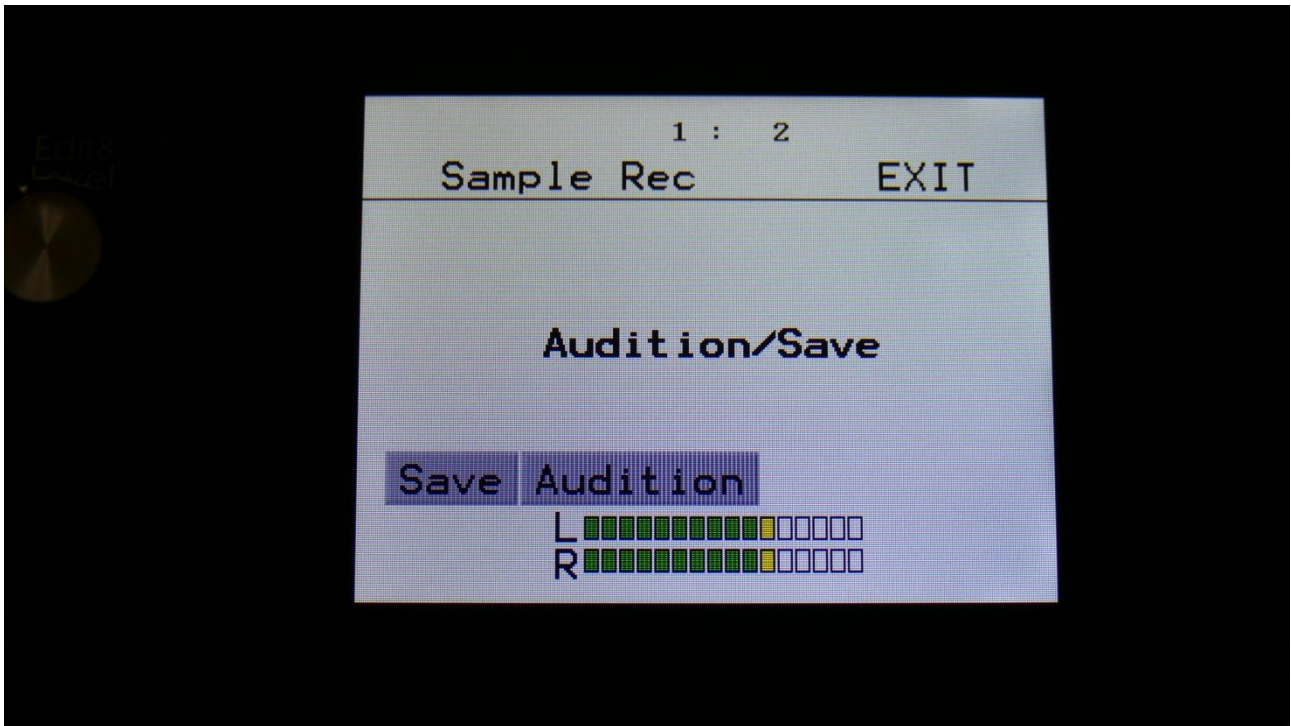
If you have the Auto parameter set to anything else than Off, it will now write: "Waiting for trigger", until a signal that is loud enough to trigger the auto offset applied.

Otherwise LD3 will now start to record the new sampling:



When you are finished recording your sample, touch the "STOP" button.

LD3 will now jump to this screen:

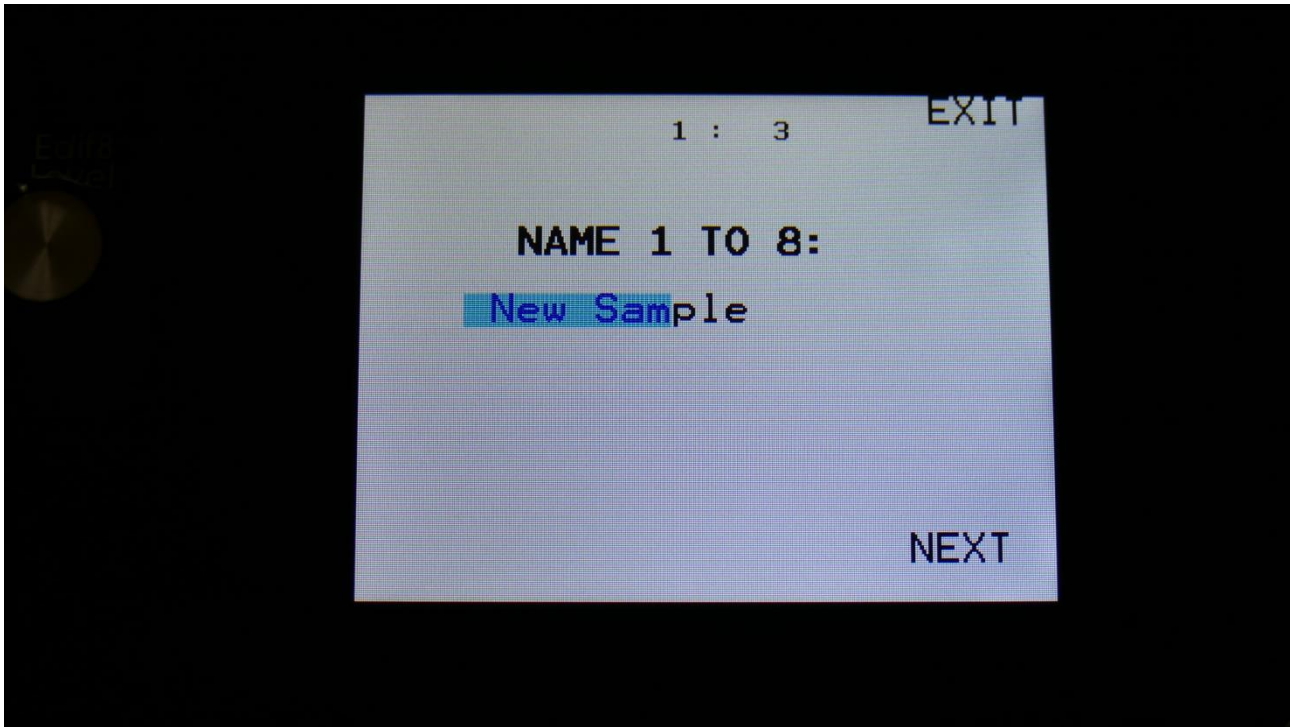


To audition the sample you have just recorded, touch the "Audition" button.

To name and save the sample, touch the "Save" button.

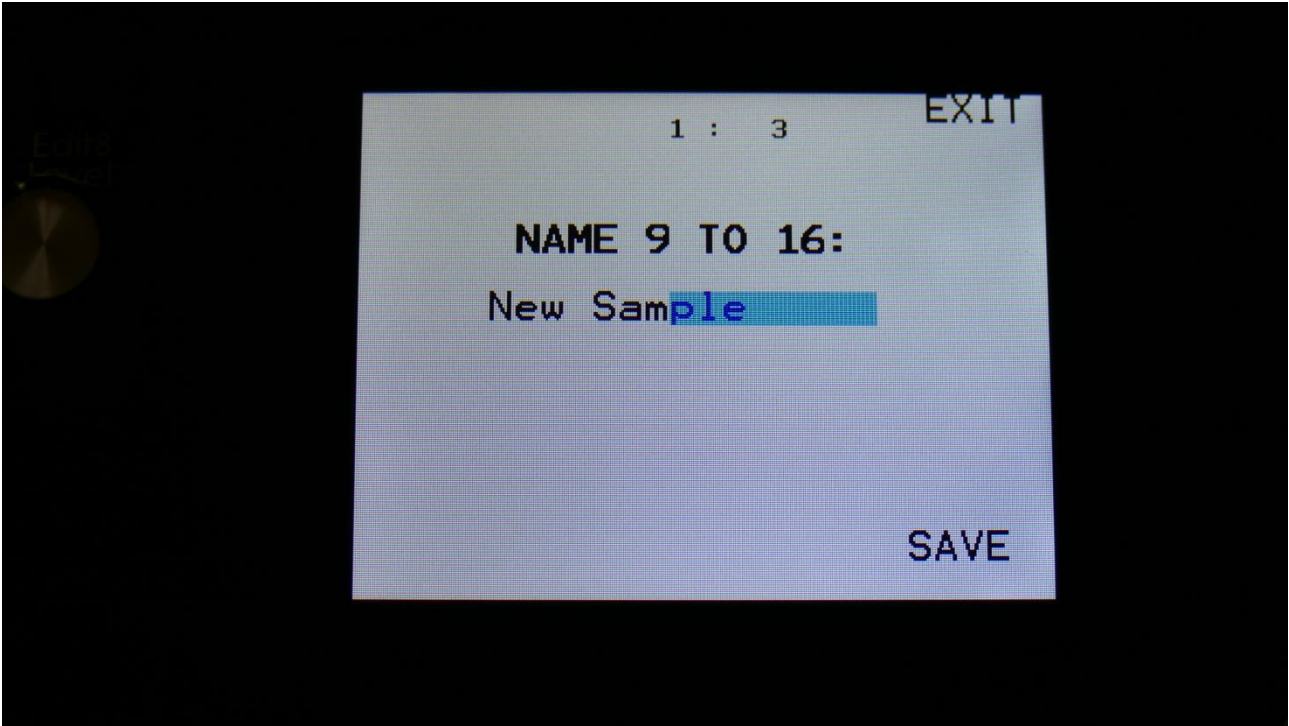
If you decide not to keep this sample, touch EXIT, and the new sample will be erased.

If you touch the "Save" button, LD3 will jump to this page:



Here you can turn edit knob 1 to 8, to select the first 8 characters of the new sample's name.

Touch NEXT to go on.



Turn Edit Knob 1 to 8, to select character 9 to 16 of the sample name.

Touch SAVE, to save the new sample, and return to the Sample Rec page.

Edit A Sampling

From the Main Synth page, touch SMP at the top navigation bar., to enter this page:



OR: Hit the Func/Mute button, so it lights up, hit step button 12 "Sample/Qrec", and hit step button 16 "Exit".

On this page you can audition samples from the FLASH memory, adjust the samples start and end points, and create sample chops.

To audition the samples that your LD3 holds, touch the **Sample** name, and the Sample Select screen will appear:



Touch A, B, C or D to select sample bank A, B, C and D, touch PREV and NEXT to view the previous or next 16 samplings.

To select a sampling, touch the sample name. To audition it, touch the Play touchbar in the bottom of this page.

When you have found the right sampling, touch OK to return to the Sample Edit page.

The Play touchbar on the sample select page re-triggers the sampling every time it is touched. If you should wish to start and stop sample playback, every time you hit play, exit to the Sample Edit page, and touch the Play button there.

Adjusting the start and end points of a sample

If a sample has silence at the start and/or end of it, or if there are parts of a sample you do not intend to use, it might be necessary to adjust the start and end points of it.

To do this, select the sample you would like to adjust.

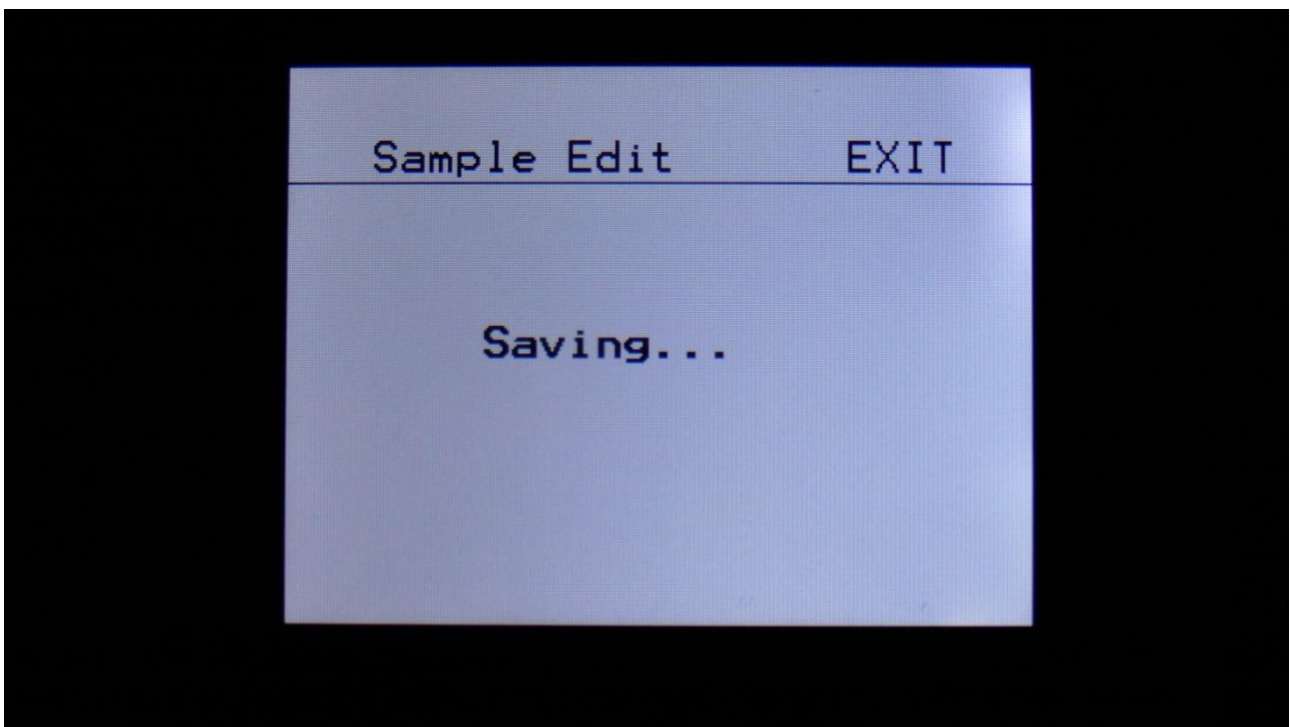
Adjust the start point by turning edit knob 1, fine adjust by turning edit knob 2.

Adjust the end point by turning edit knob 3, fine adjust by turning edit knob 4.

Audition by touching the PLAY button.

When you are satisfied with the result, touch the SAVE button.

LD3 will now show:



Instead of searching manually for the sample start point, it is possible to use the Chop function for this. This is described in the “Sample Chops” section, that starts on the next page.

Sample Chops

On LD3 it is possible to add chop points to a sampling, in order to make it play back a certain portion of the sampling at a time. Sample chops are generated in the Sample Edit section, to be used by the samplers in the Synth section.

This function could be used for the classical separating single drum hits from a beat, or to find the startpoint of a sampling in a fast way, if the chop function is set up for peak detection.

The chop function of LD3 isn't though limited to this. On LD3 it is also possible to make the chop function find "wave chops". This function will pick out single wave cycles of the sampling, and create chop points for these. In the samplers of the synth section, it is then possible to select these wave cycles, loop them, and switch between other wave cycles of the sampling. This can also be used as a super easy way for looping a sampling.

All chop points are non-destructive to the sampling. Only the positions of the chop points are stored, nothing is changed on the sampling itself. Chop points can, at any time, be moved, deleted and inserted.

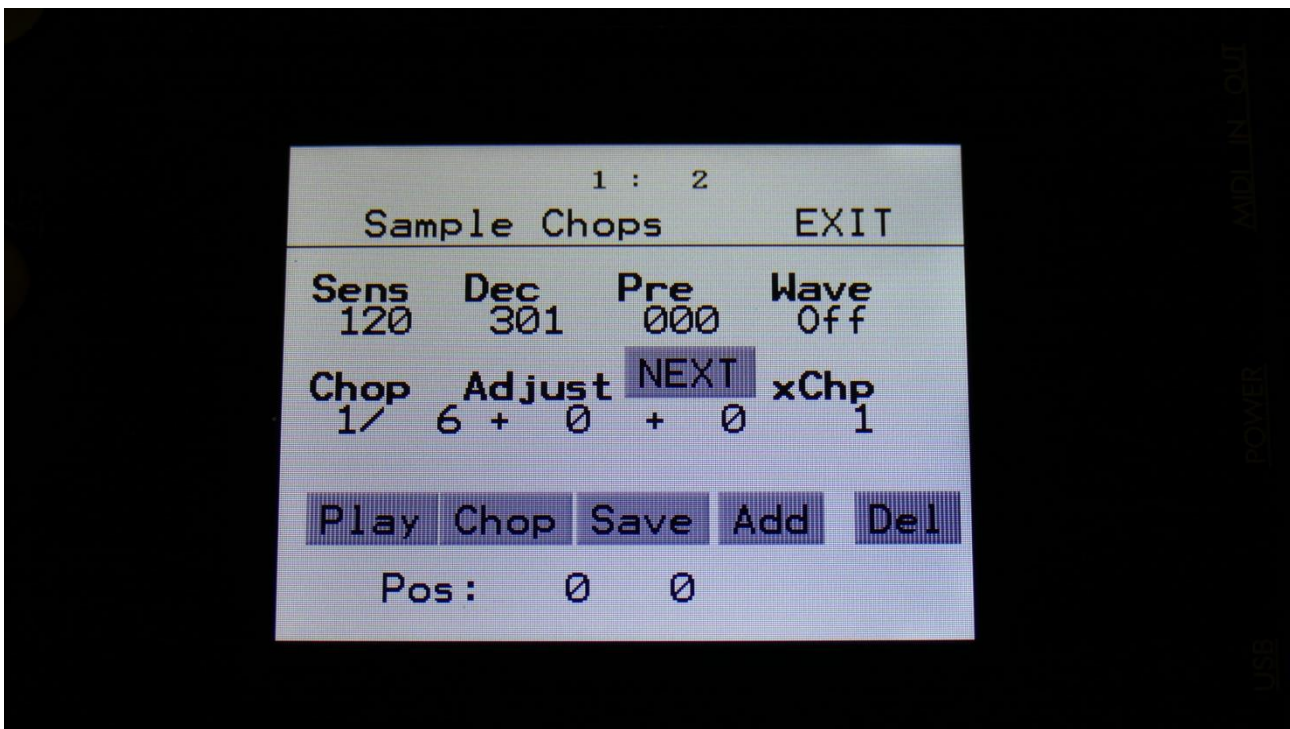
Up to 64 chop points can be created for each sampling.

Another way to create chop point, are to use the "Import Samplings From Dir As Chop Points", described in the USB section of this manual.

If you have already created chop points on a sampling, in a computer program that uses cue points, like FL Studio, or any other gear that generates cue points, LD3 will import these as chop/loop points. When you exports LD3 samplings via USB, any generated chop points will also be exported as cue points, for use in other gear or computer programs.

Remember to save the chop points. Else they will be lost!

To create and/or edit sample chop points, from the Sample Edit page, touch the **Chop** touchbutton, to enter this page:



The parameters of this page:

Sens: Adjust how loud the level of the sample audio should be, before it detects a chop point. If it creates too many chop points, turn this parameter up a bit. If it creates too little or no chop points, lower this parameter.

Dec: Chop envelope decay. This adjusts the decay time of the chop detection envelope in a reverse manner. The lower the value, the slower the decay. If the sounds is recorded with very little or no silence between them, like in a beat, this needs to be set to a high value. If there's a good gap of silence between the sounds, set it to a low value.

Pre: If there are loud clicks in the start of many of the chop points, especially on low frequency sounds, turning up this parameter will make LD3 set the chop points a little bit earlier, than it usually would. This will remove these clicks.

Wave Mode: Switches the chop wave cycle mode on and off.

Chop: Select a chop number for auditioning. Total number of chops is shown right after the /.

Adjust: (2 parameters). For adjusting the selected chop forward (positive values) or backwards (negative values).

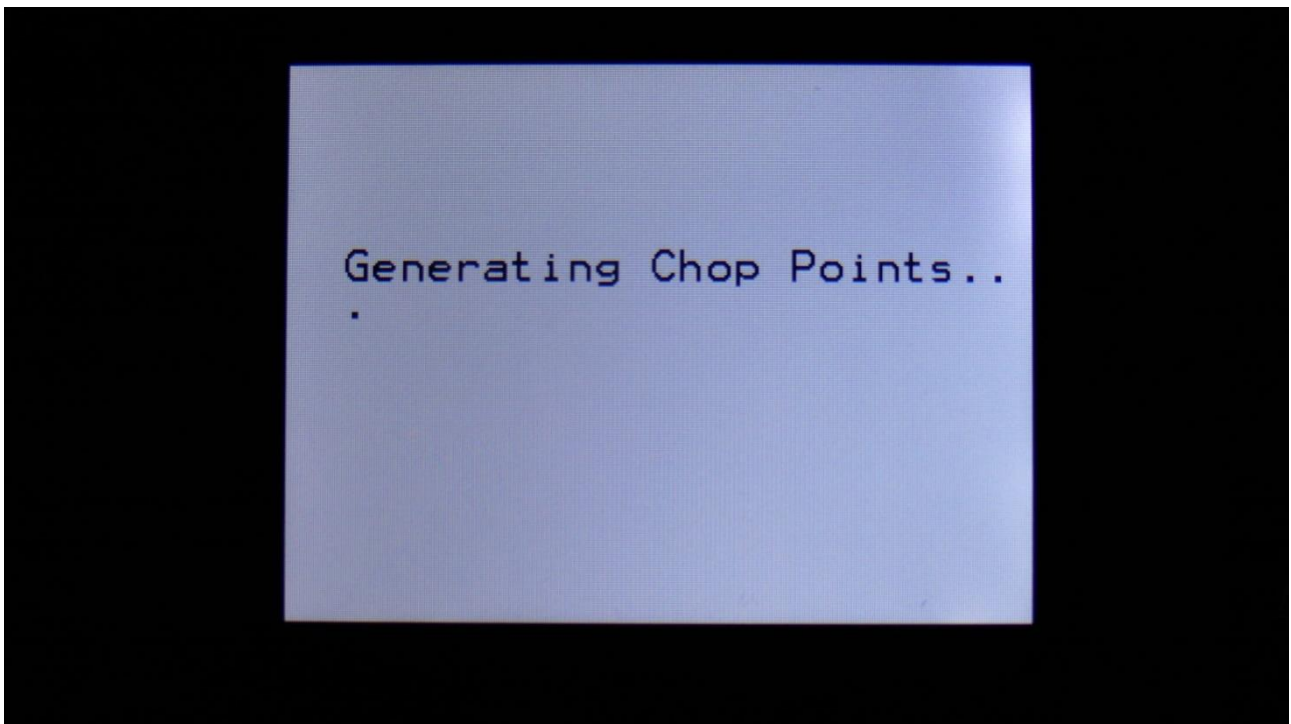
xChp: Number of chops in each chop point. Turning this up, before generating chops, will make each chop contain a number of chops. Especially usefull, when using the wave chop function for making sample loop points.

Generating Sample Chop Points

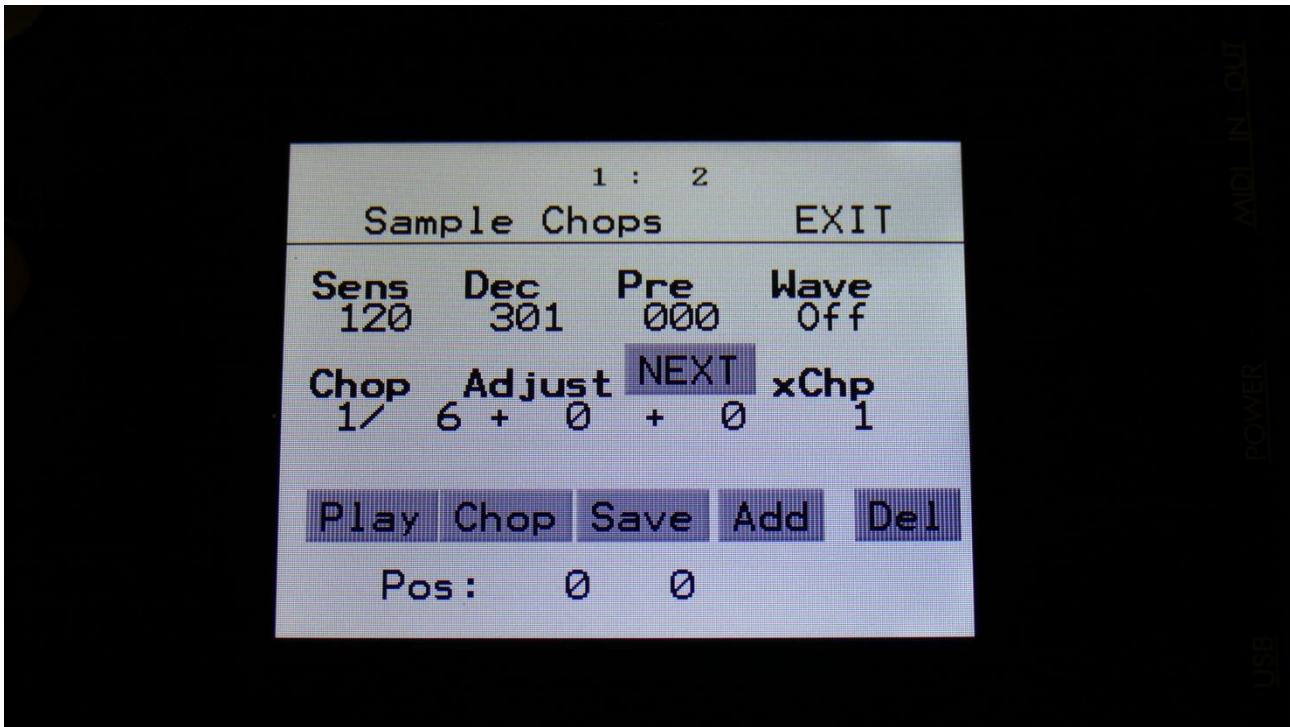
On the Sample Chops page, adjust the Sens, Dec, Pre and xChp parameters as desired, and switch wave mode on if desired.

Touch the **Chop** button.

LD3 will now look for chop points, in the selected sampling:



When it is done, it will jump back to the Sample Chops screen:



By adjusting the Chop parameter, it is now possible to select and audition the chop points that LD3 has found.

Audition the chop points by hitting the “Play” button. If some chop points are not exactly where you would like them to be, adjust them with the Adjust parameter. If some chop points shouldn’t be there, delete them by hitting the “Del” button.

It is also possible to adjust the end point of the chop point. Simply touch the **NEXT** button. Now, while auditioning the selected chop point, you can adjust the next chop point, which is also the end point for the selected chop point, using the Adjust parameters.

To add chop points, first touch the “Play” button, to make the sample/chop point play back. While the sampling is playing back, hit the “Add” button when it is playing back the point, where you would like to add a chop point.

In the bottom of the screen, it shows the position of the selected chop, in the values that the **Sample Start** parameters would have to be adjusted to, to select this point of the sampling. So if you don’t want to use the chop points anyway, but just wants the sample playback to start at this point, you can exit to the Sample Rec page, and set the Sample Start point to this value.

When you are satisfied with the chop points, remember to hit the “Save” button to keep them. The chop points are saved together with the sample data, and are also imported/exported via USB as cue points.

Wave Chop System

This system can take any of the samplings held in your LD3, and chop it up in, up to 64, single cycle waves.

If you already have a library of single cycle waves in .wav format, you can use the “Import Samples From Dir As Chops” function, to import the single cycle waves into one sampling with wave-chops.

When you have a sampling with the wavechops added, it is possible, in real time, to select which single cycle should play back, in the samplers in the synth section, using the “Chop” parameter. It is also possible to select single cycle waves, using any modulation source.

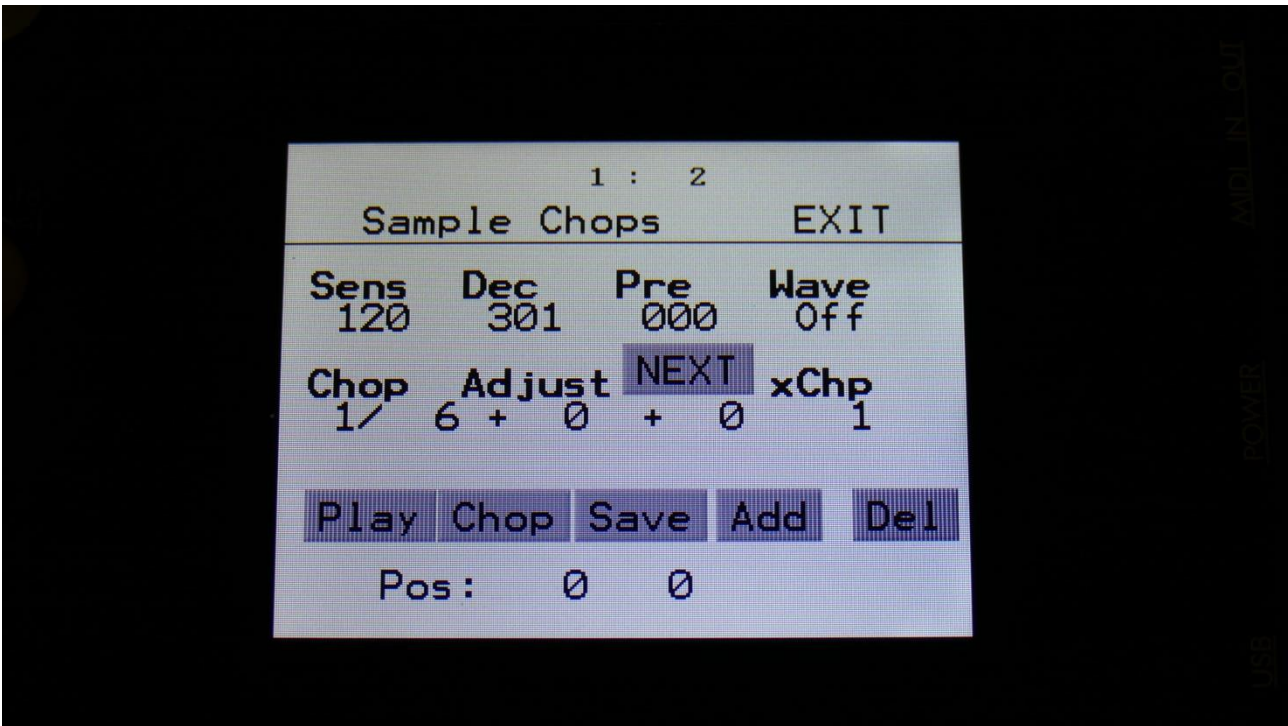
This gives some really cool possibilities, like wave sequencing, wave tables and time stretch.

How to wave chop a sampling:



On the sample Edit page, select the sampling, you would like to wave chop, and touch “Chop”.

That will get you to the Chop page:



Set the Wave parameter to On.

When in Wave chop mode, the Dec parameter functions in the opposite way, as when in normal chop mode. The lower the value, the faster the decay.

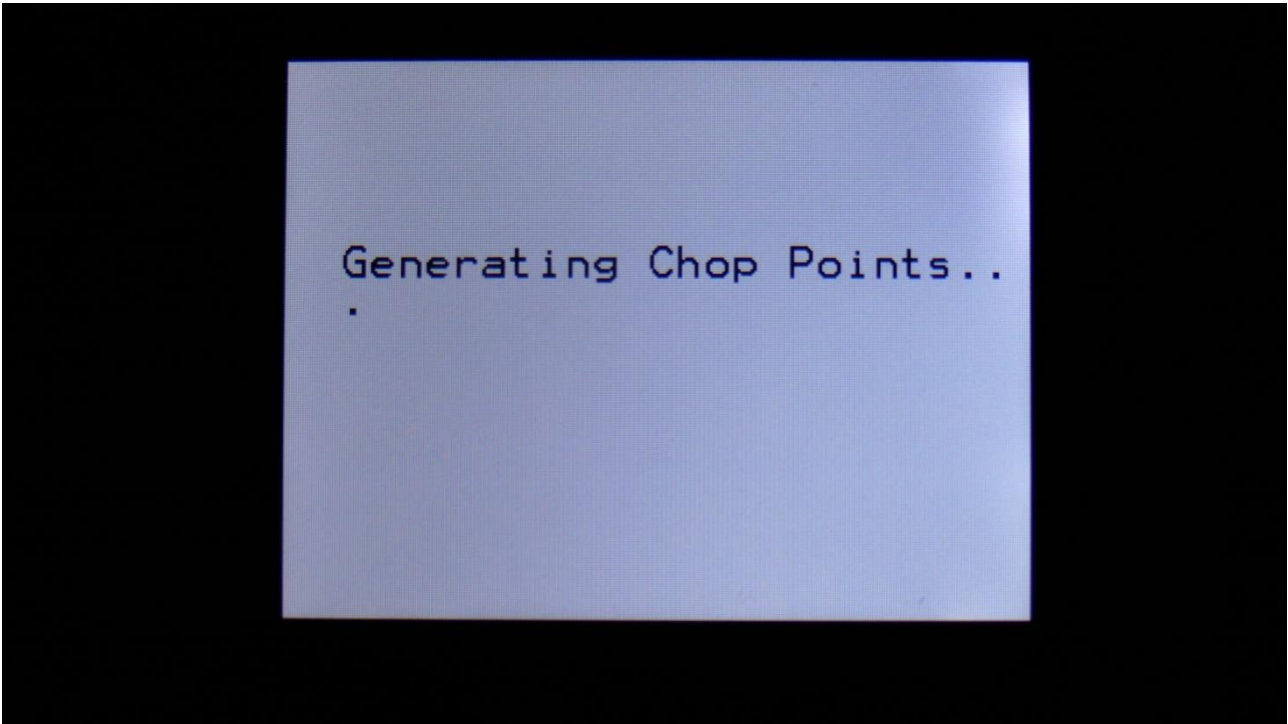
A good startpoint for wave chopping is to set Dec to 1 and Sens to 128. If the wave chops gets too long (very few chop points), try first to adjust Sens down. If this makes things worse, adjust it up. On simple waveforms, Sens should, in most cases, have a low value. On more complex waveforms, Sens should have a higher value.

If you wish your wavechops to be larger than single cycles, you can either turn Dec or xChp up.

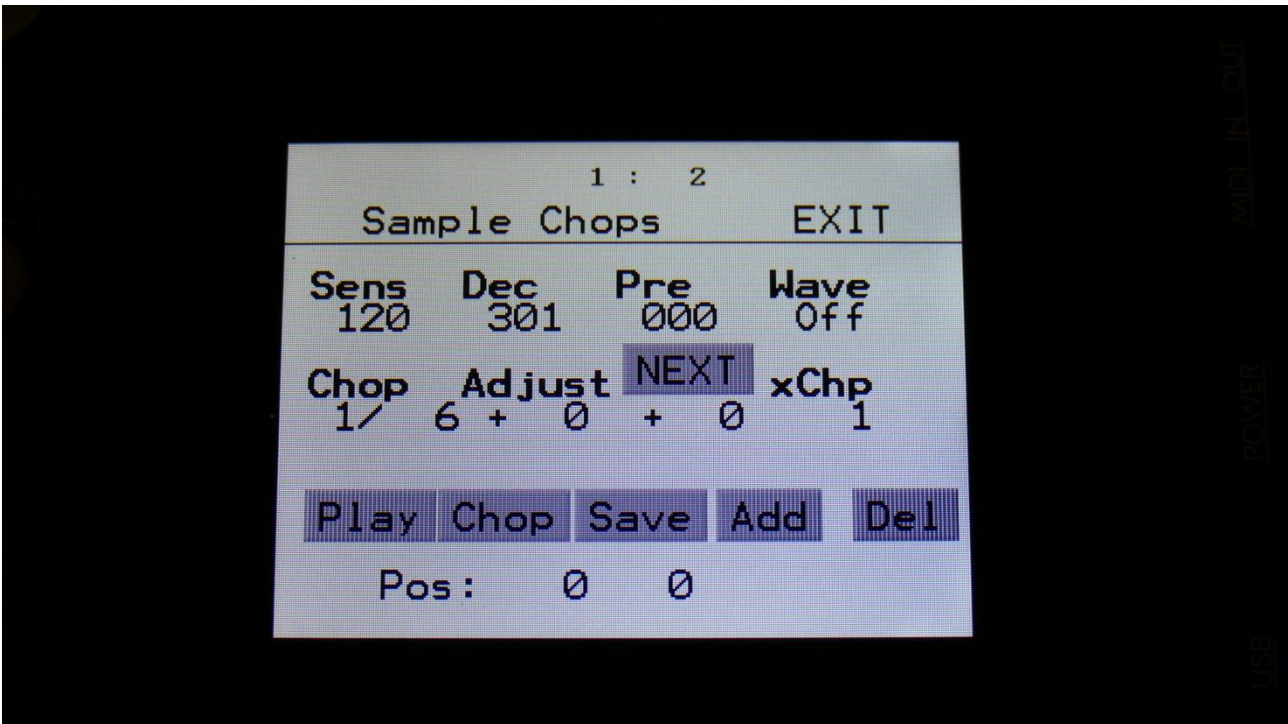
The Pre parameter should be kept at zero.

When you have switched Wave on, and adjusted the settings, touch the "Chop" button.

LD3 will now show:



When it has found some chop points, it will return to the Chop page:



Now touch the play button, and select chops by adjusting the Chop parameter, to audition the result.

If you are not satisfied with the result, re-adjust Sens, and hit Chop again.

When you have found a result, that you are satisfied with, please remember to touch the “Save” button.

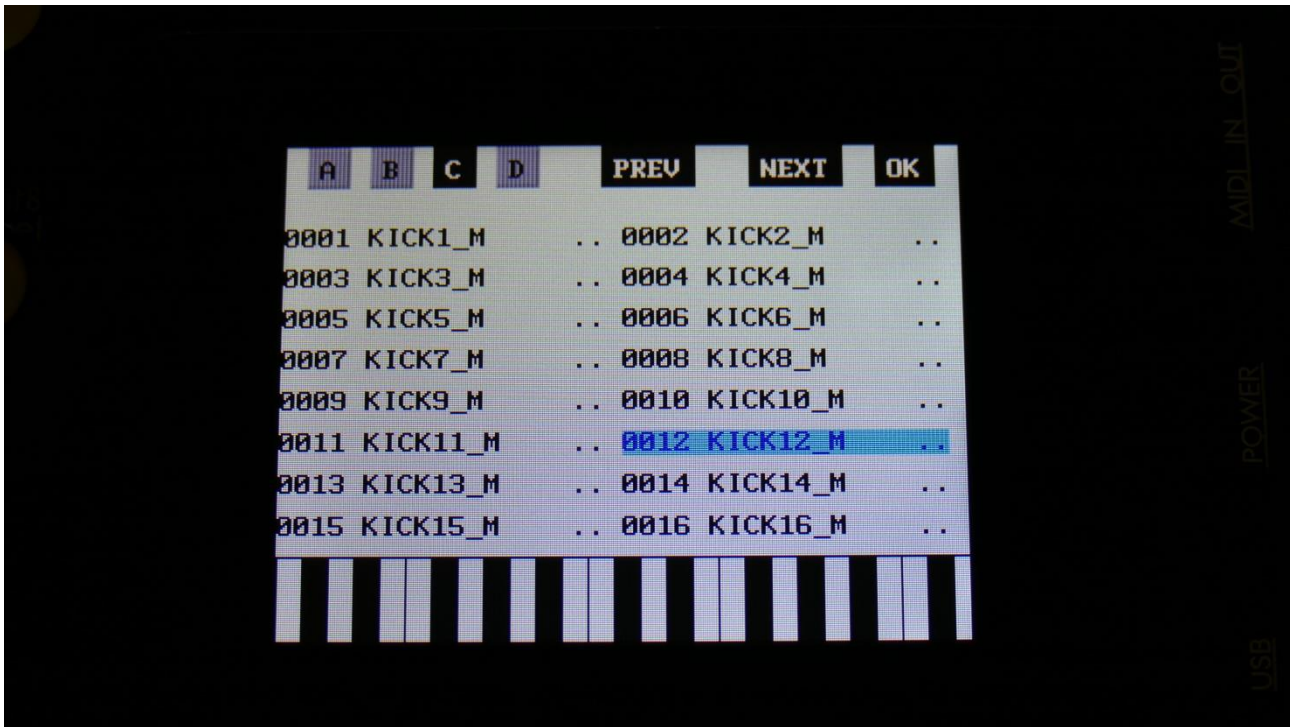
To use the wave chops for something, now exit Sample Edit, and go to a part Osc page.



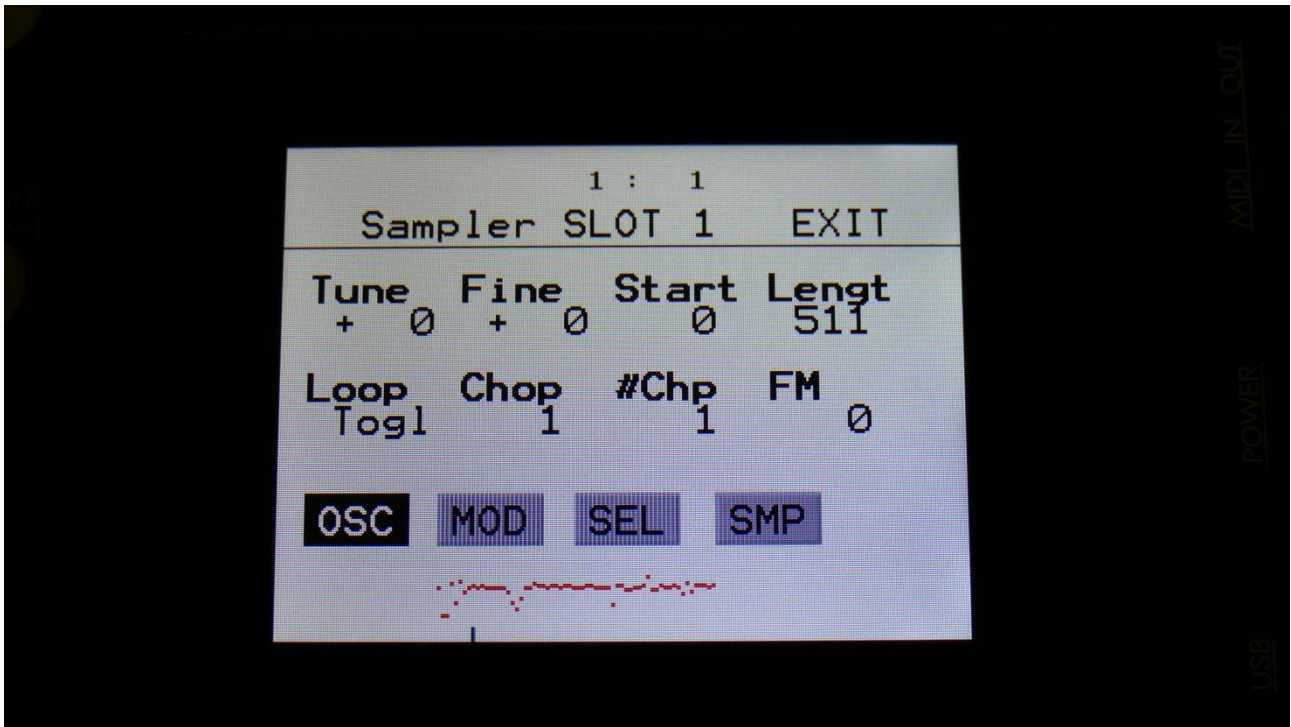
Set Osc1 in sampler mode, and touch the SMP touch button to select the sampling, that you just wave chopped (or any other sampling that you have wave chopped).



Touch the sample slot 1 name, to open the sample select page.



Select the sampling, touch OK, and touch the OSC touchbutton.



Set Loop mode to On, and adjust the Chop parameter, to select the single cycle waves. If you would like more single cycle waves to play back inside the loop, turn the #Chp parameter up.

Modulating the Chop parameter



Unlike most other parameters in LD3, only positive modulation are applied to the Chop parameter. So the Chop parameter selects the lowest possible chop to be played back.

Time Stretch

To time stretch a wave chopped sampling:

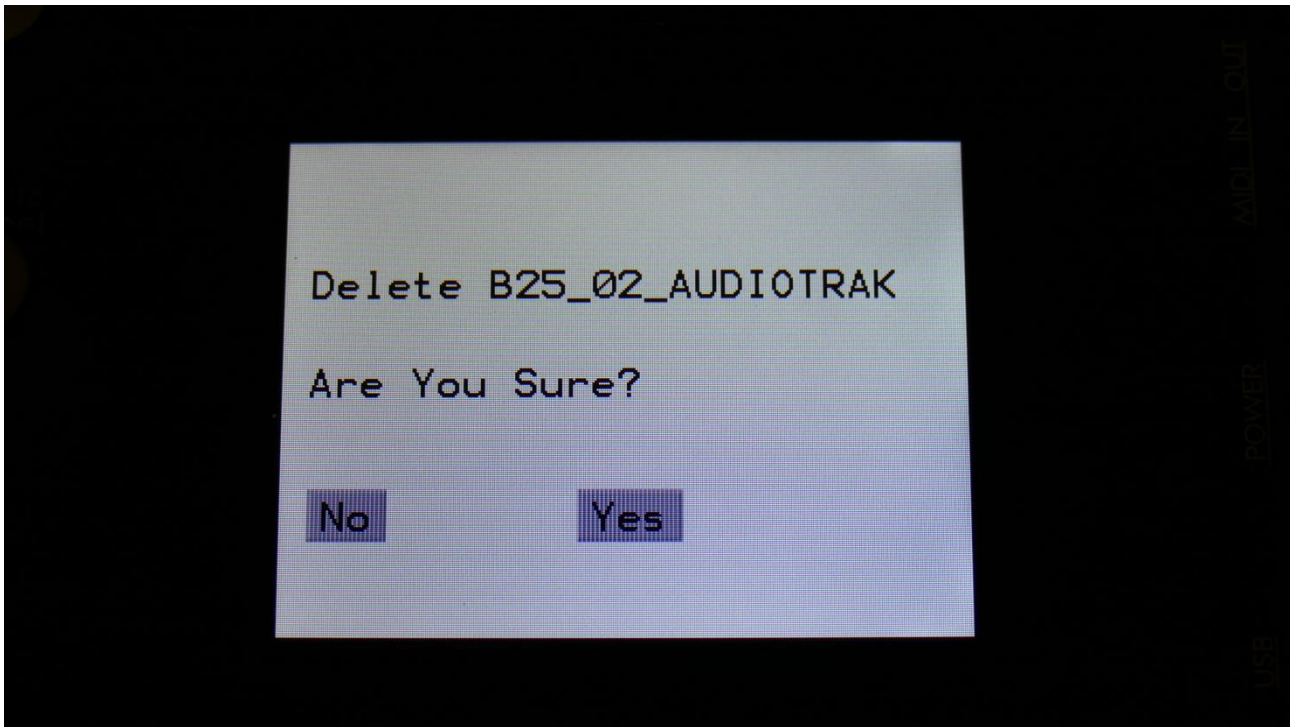
- Set the Chop parameter to 1.
- Set up Envelope 1- (Envelope 1 inverted) to modulate the Chop parameter, and turn the modulation amount up. You can obtain different effects, with different values of the modulation amount.
- Adjust the Envelope decay, to make the playback of the sampling faster (low decay values) or slower (high decay values).
- Adjust the attack parameter, to obtain a reverse playback effect, in the start of the sample playback.
- Modulate the envelope parameters, to have dynamic sample lengths.
- The sampling will play back at the exact same speed, over the entire keyboard range, regardless of the sample pitch.

Deleting last recorded sampling

On the Sample Delete page, it is possible to delete the last recorded sampling in sample Bank A, B, C or D. On the sample select page, select the sample bank, from which you would like to delete a sampling.



Touch DEL.



To delete the sample, touch "Yes".

If you regret, touch "No", to return to the Sample Edit page.

Deleting other samplings, than the last recorded one

Because of some security features of the flash memory used in LD3, it is only possible to directly delete the last recorded sampling in each bank.

By using its USB functions (explained in the next chapter of this manual), it is though possible to delete other samplings.

To delete another sampling:

- Connect a USB drive, that you have tested to be working (try to import some samplings from it), to the LD3 USB port.
- Create a new directory.
- Export all your samplings of the bank, where the sampling(s) you want to delete is located, to the USB drive, inside the new directory.
- Enter the directory, on the USB drive, and select the sampling(s), that you want to delete.
- Push "MD/del" on the USB page.
- Push "Del".
- Confirm.
- Leave the USB pages, go to the "More..." section, and delete the whole sample bank.
- Go back to the USB page, and select the directory, holding your samplings.
- Push "Import", and then "Re-Load".
- LD3 will now put your samplings back on the right place, except for the sampling(s), that you deleted.

Graphical sample editing



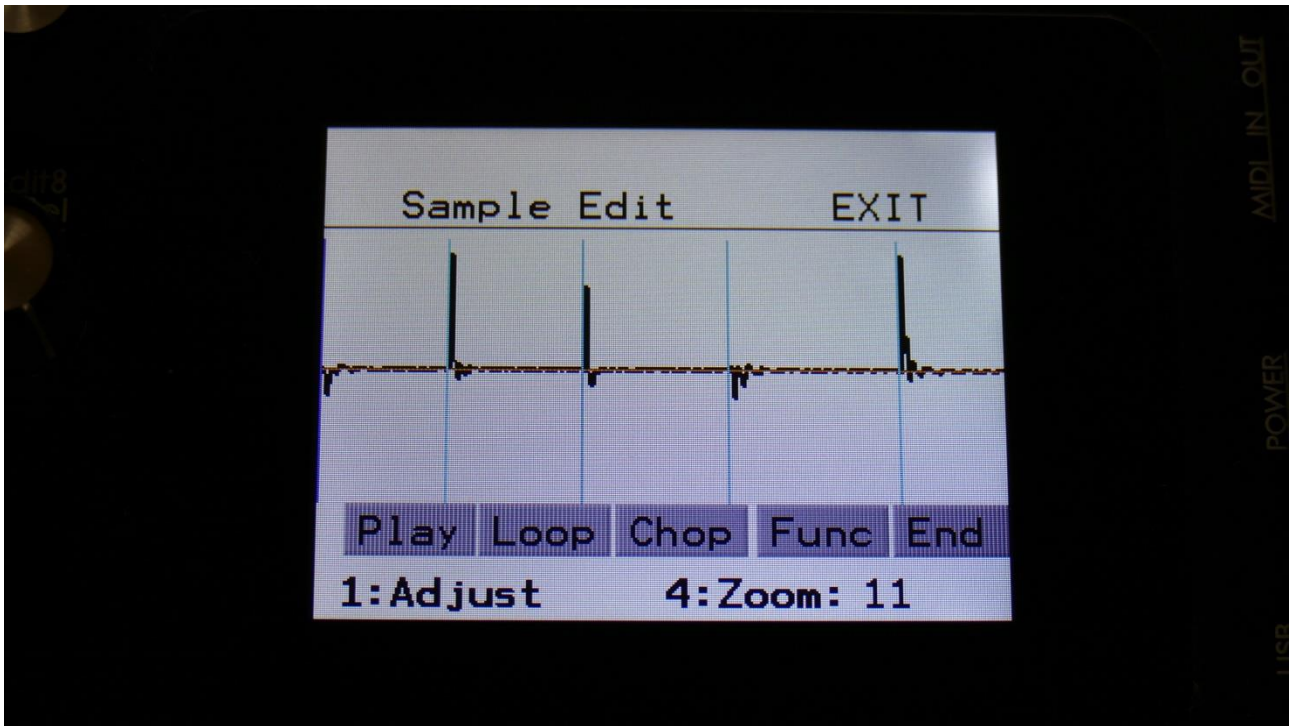
On the Sample Edit page, the touch button previously named “Func”, has been renamed to “Grap”.

Touching this will enter the graphical sample edit pages.

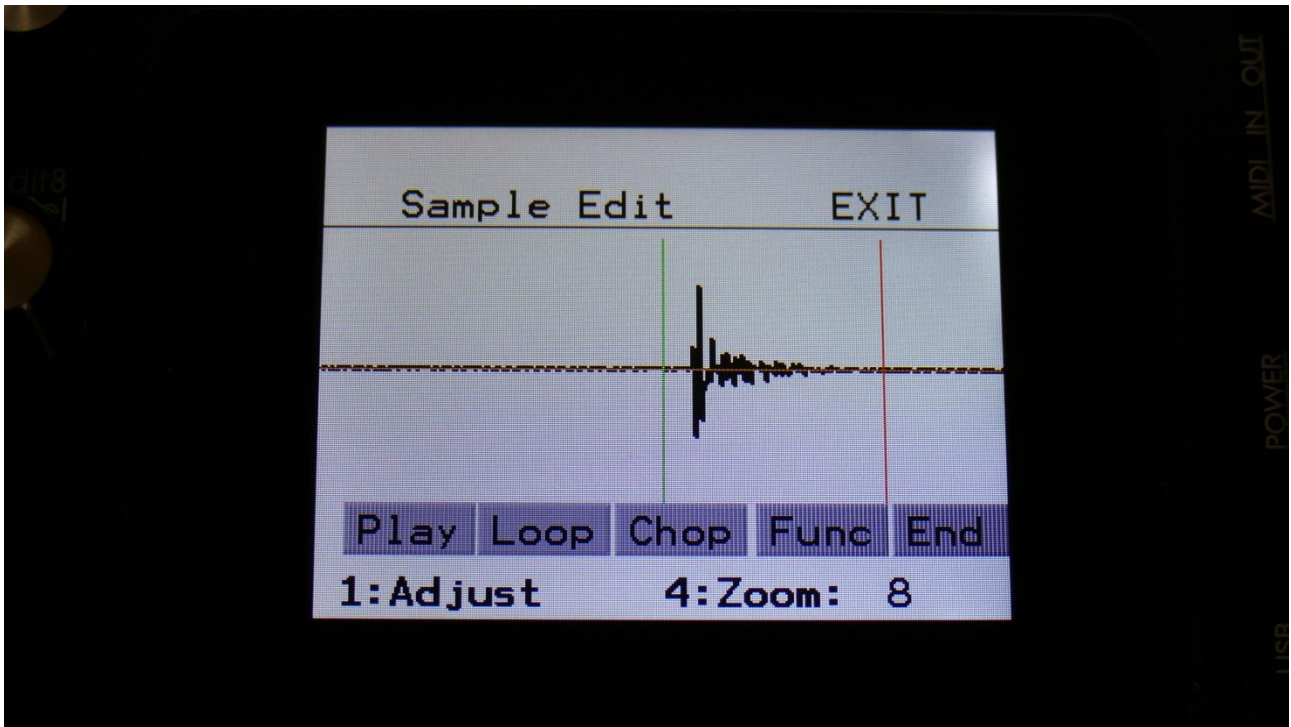
On the graphical sample edit pages you can add and remove chop/loop points and select a portion of a sampling, to be saved as a new sampling.

It has also been made easier, to set loop points in long samplings, by making it possible, to save the selected end point as a chop/loop point.

From the Sample Edit page touch Grap, to enter the graphical edit page:

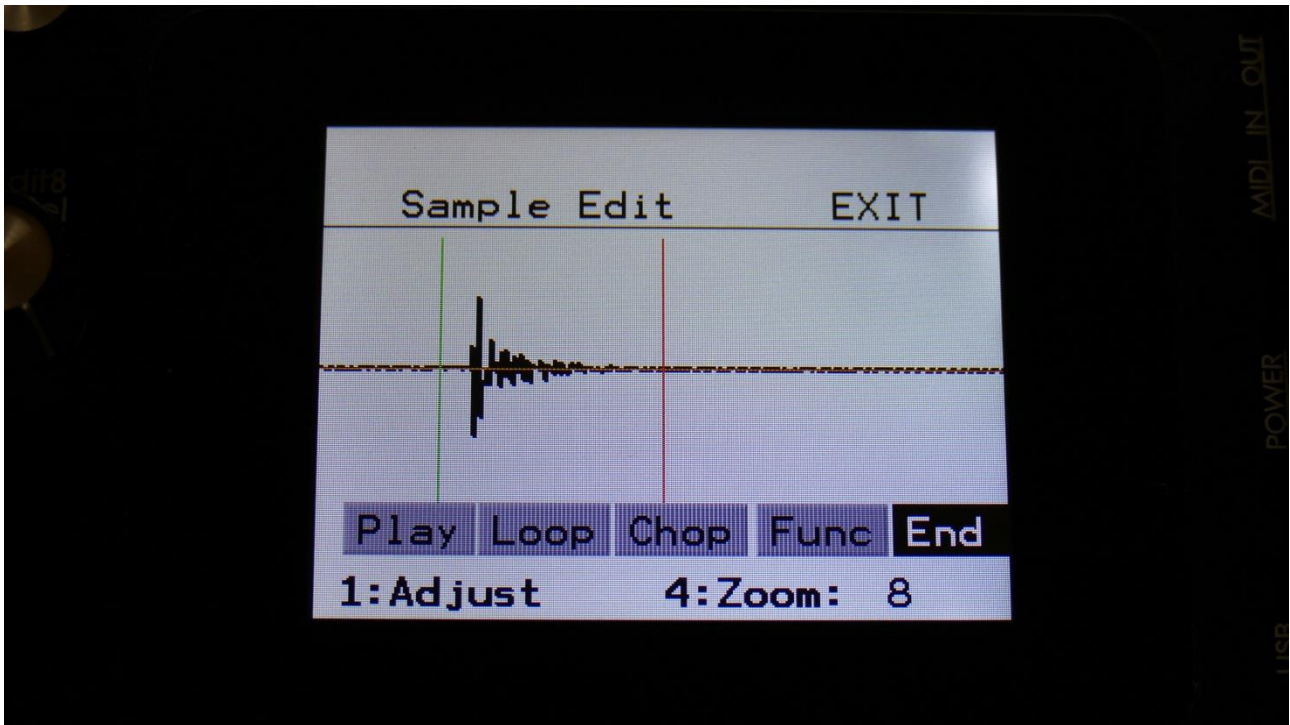


On this page, you will see a graphical presentation of the selected sampling. Chop points are indicated as vertical blue lines, and the sample zero point is indicated as a horizontal red line.



By touching anywhere on the screen, or by rotating Edit Knob 1, you can select a start point. This is indicated by a vertical green line.

By rotating Edit Knob 4, you can zoom in and out.

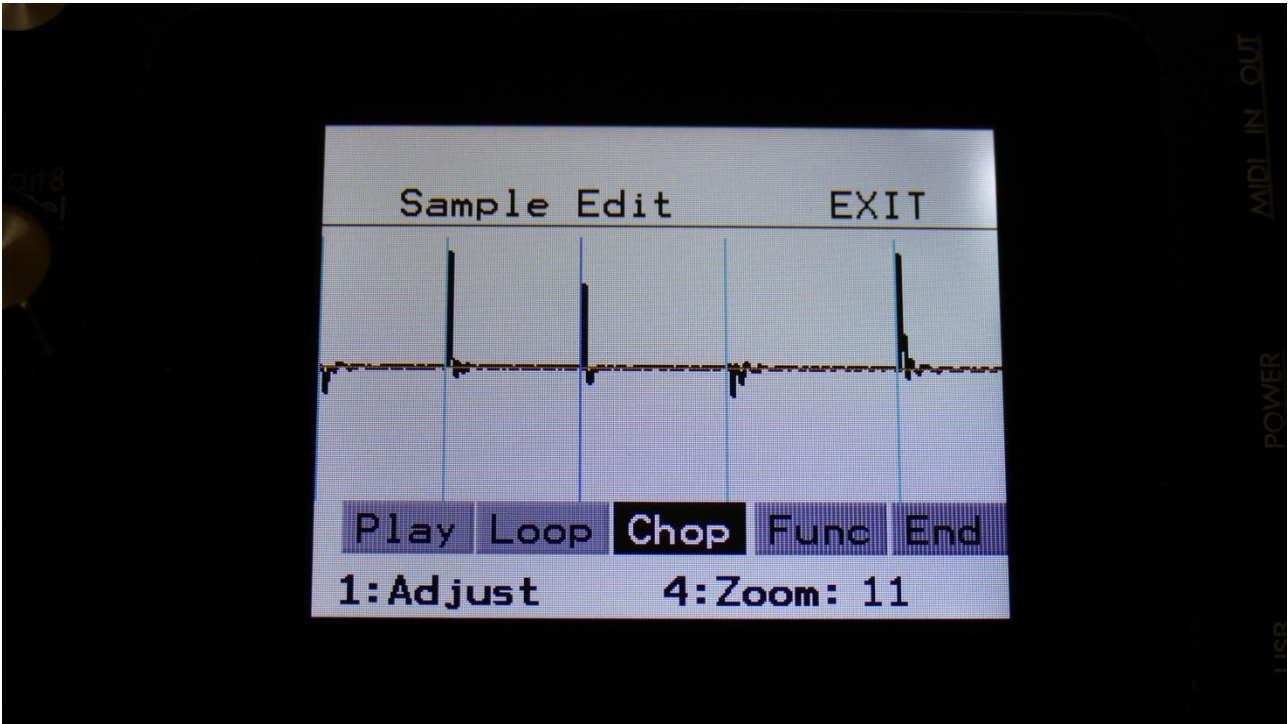


By touching the End touch button, so that it turns black, the screen will now be focused on the end point, which is indicated as a vertical red line. When you touch the screen now, the red line will move, indicating that the end point is selected. It is also possible to change the end point, by rotating Edit Knob 1.

If you touch the Play touch button, the sample will play back from the start point to the endpoint, and then stop.

If you touch the Loop button, so that it turns black, and then touch the Play button, the sample will play back from the start point to the end point, and then loop back to the start point, and play back the selected piece again, until Play or Loop is touched again, so that any of these turns grey.

Selecting and edit chop points



Touch the Chop touch button, so that this becomes black.

Now, when you touch the screen, you will select a chop point. The line indicating the chop point will turn dark blue, when it is selected.

Rotate Edit Knob 1 to adjust the chop point. Rotate Edit Knob 4 to zoom in and out.

If you touch the Play touch button, the sample will play back from the selected chop point to the next chop point, and then stop.

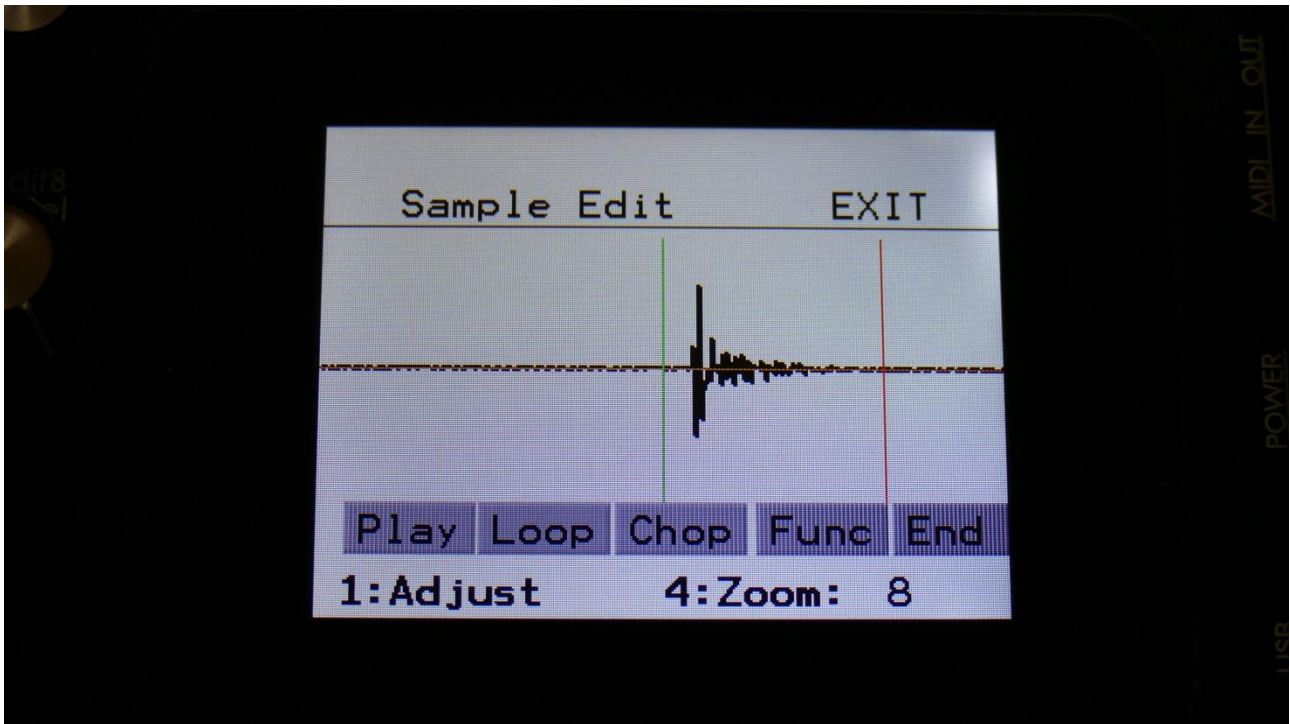
If you touch the Loop button, so that it turns black, and then touch the Play button, the sample will play back from the selected chop point to the next chop point, and then loop back to the selected chop point, and play back the selected piece again, until Play or Loop is touched again, so that any of these turns grey.

By touching the End touch button, so that it turns black, it is possible to adjust the next chop point, which is also the end point of the selected chop, by rotating Edit Knob 1.

Adding a chop point

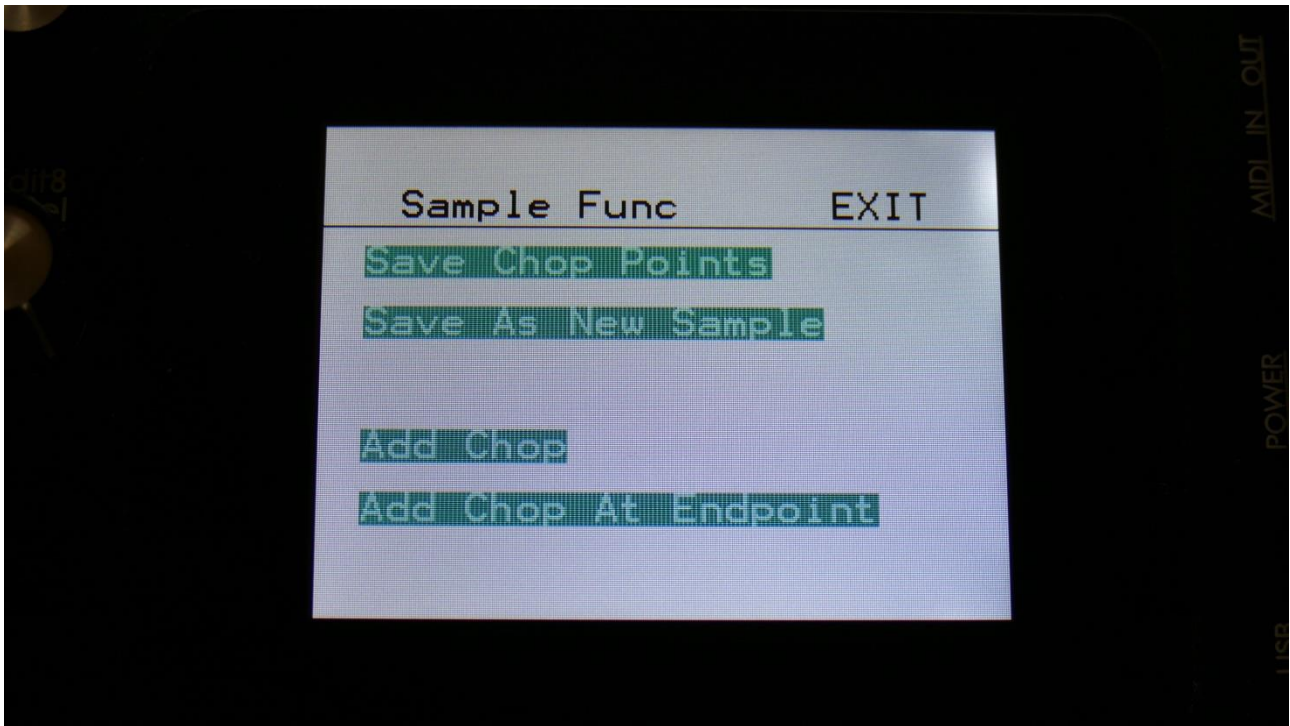
Make sure that the Chop touch button is grey.

Set the start point, where you want the chop point to be, either by touching the screen, or by rotating Edit Knob 1.



Touch the Func touch button.

Now you will enter the graphical function page:



Touch "Add Chop". LD3 will now jump back to the graphical edit page, and a chop point will be added at the selected start point.

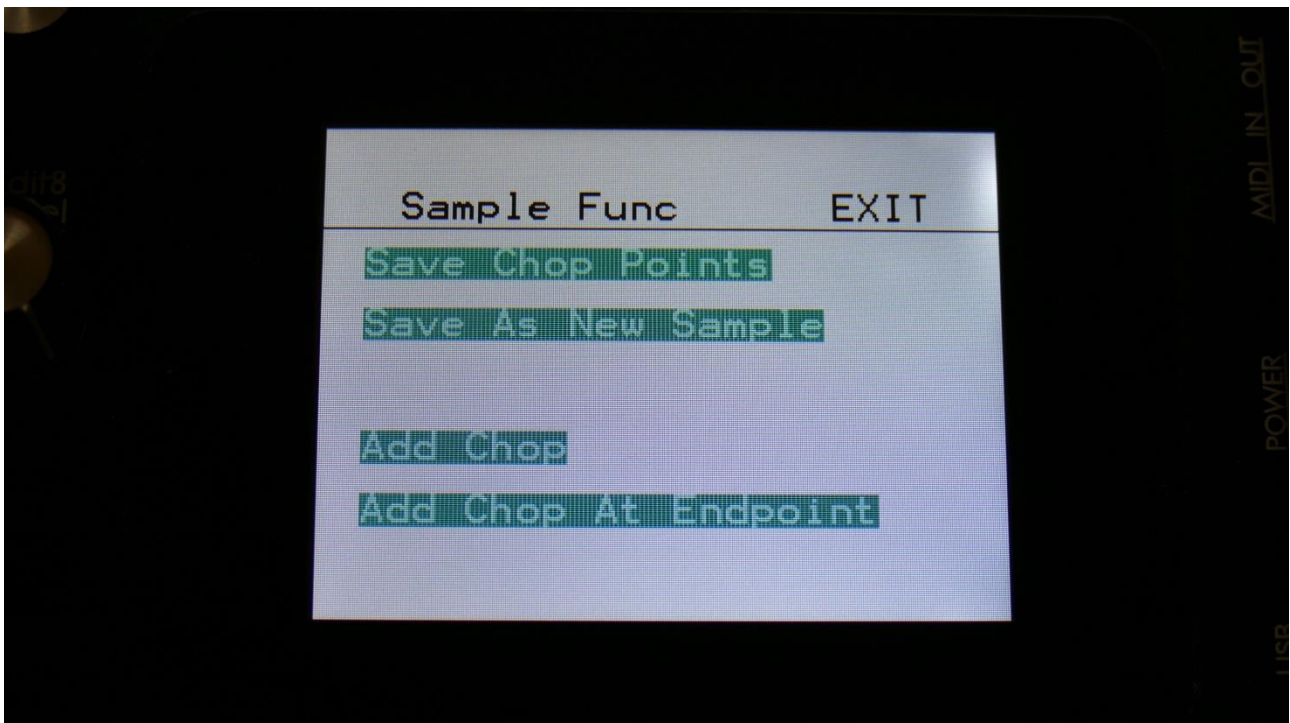
Adding a chop point at the selected end point

If you wish to create a loop, for instance for use with the sustained loop feature, you will adjust the start and endpoints, until the loop sounds as intended.

Then you will save the start point as a chop point.

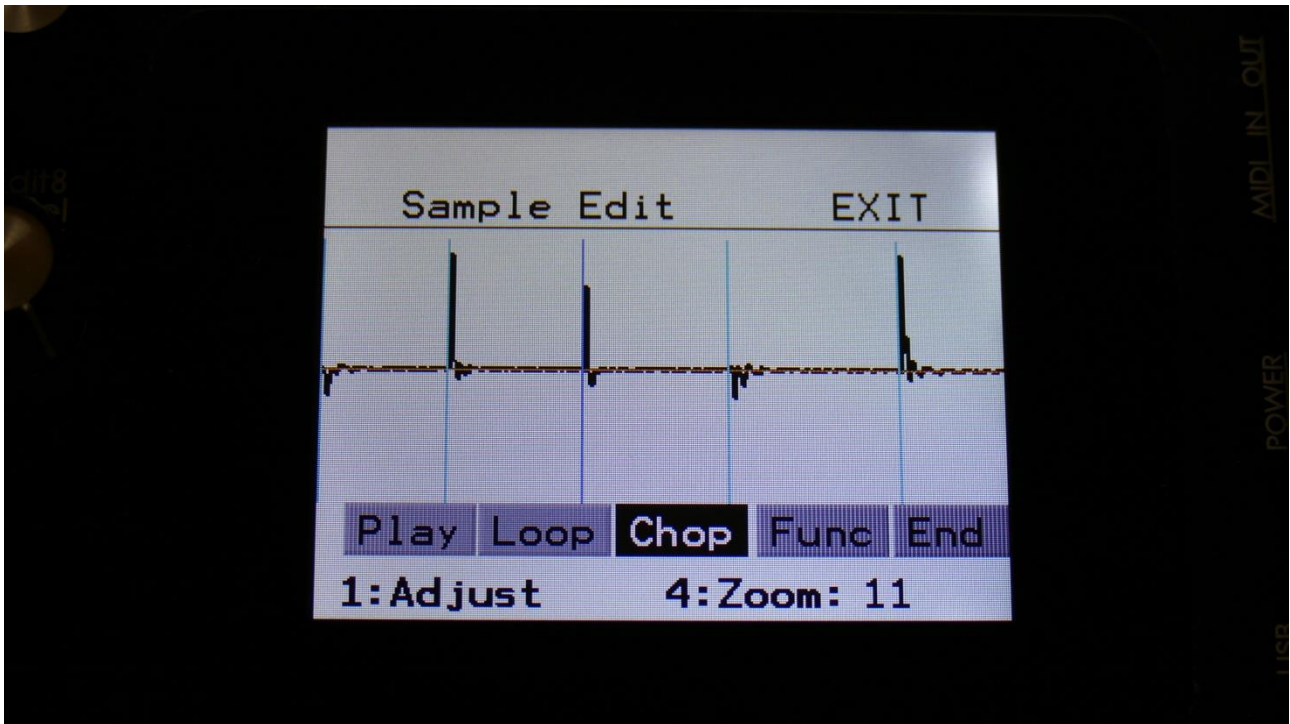
In order to make the loop work, you will though also need to save the selected end point as a chop point.

To do this, you must enter the graphical function page again:



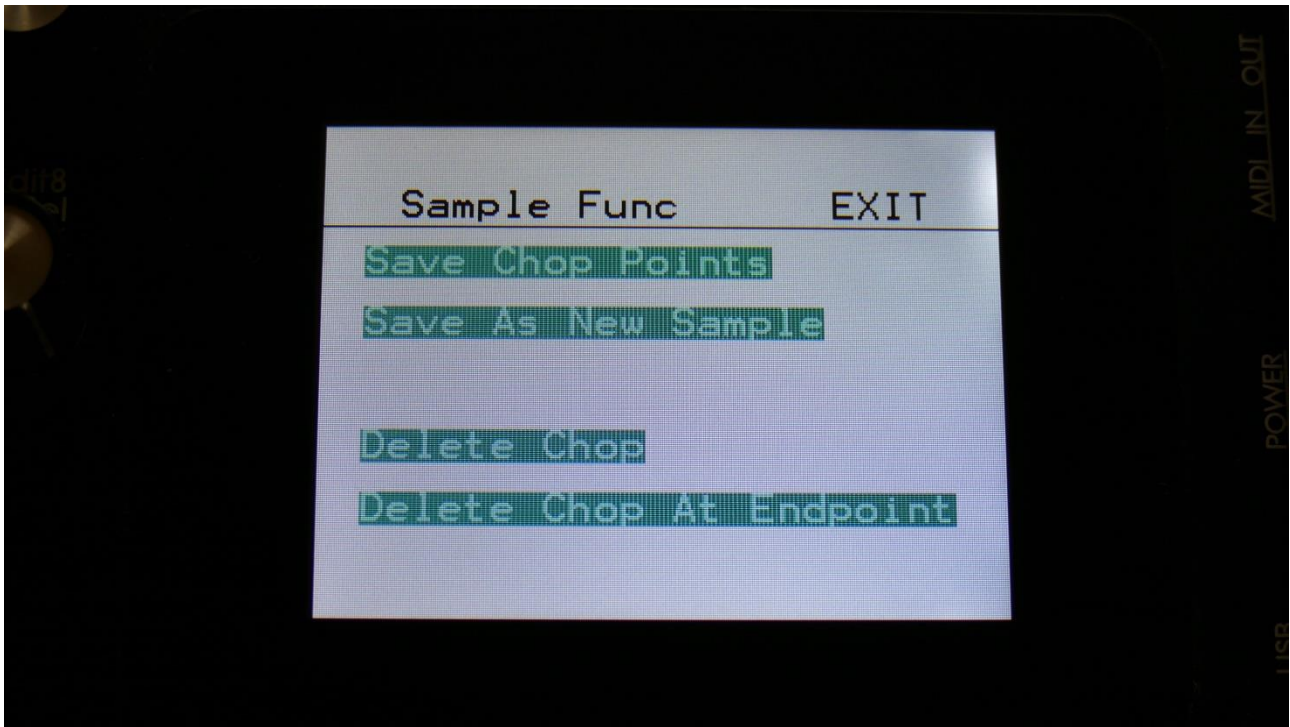
Touch "Add Chop At Endpoint". LD3 will now jump back to the graphical edit page, and a chop point will be added at the selected end point.

Delete a chop point



Make sure that the Chop touch button is black, and that you have selected the chop point, that you want to delete.

Touch the Func touch button, to enter the graphical functions page.



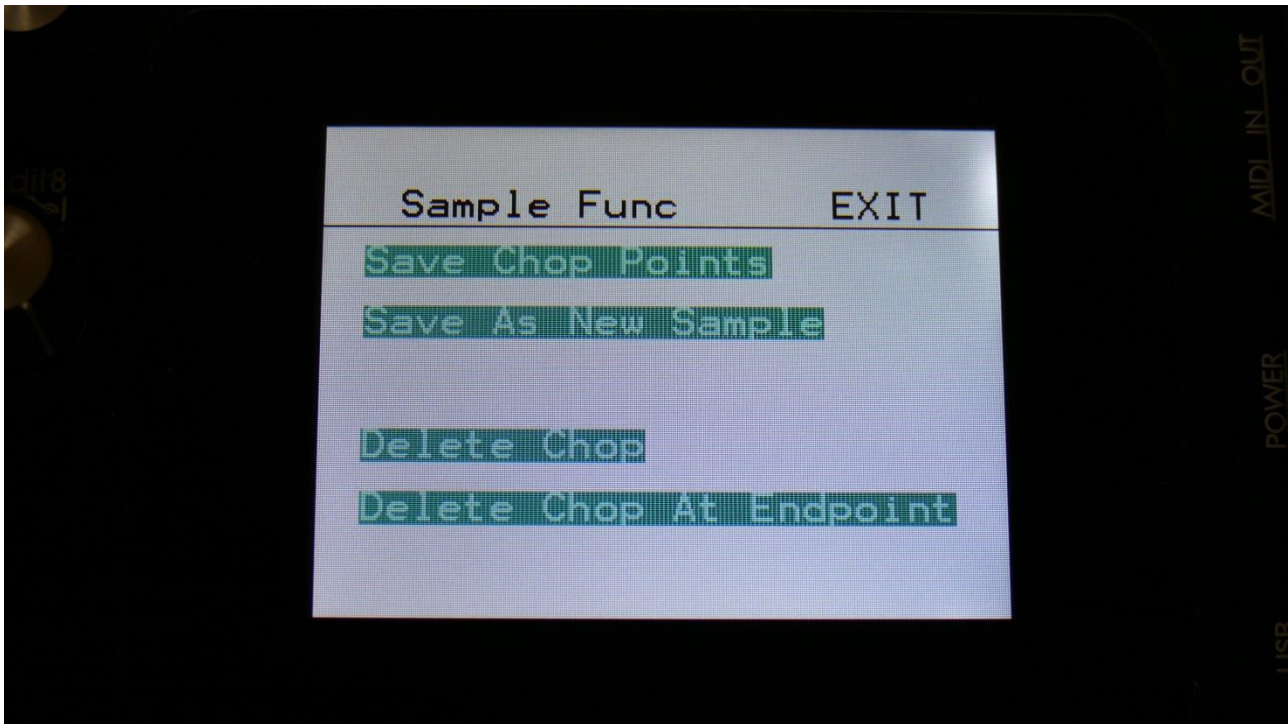
Now touch "Delete Chop". LD3 will now jump back to the graphical edit page, and the previously selected chop point will have been removed.

To delete the next chop point, follow the same procedure, but touch "Delete Chop At Endpoint" on the functions page instead.

Saving the chop points

After you have edited, added and deleted chop points, you must save these changes, in order to keep them.

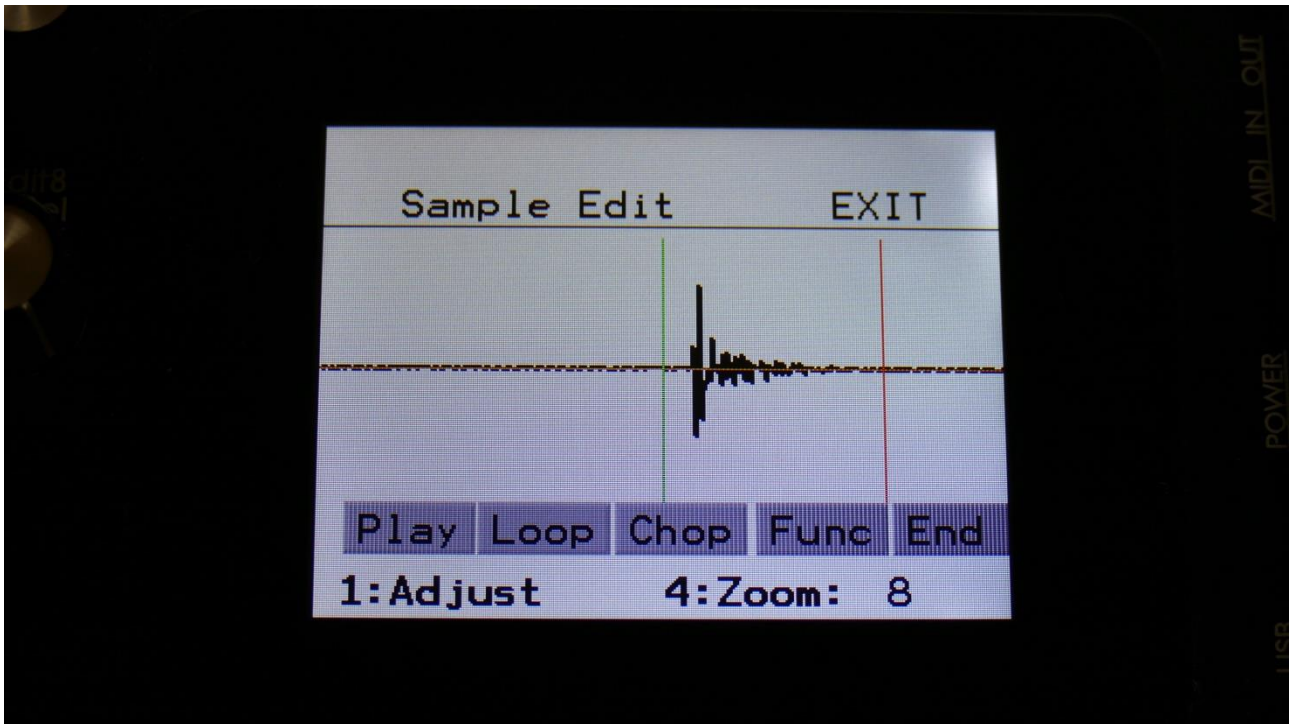
This is also done from the graphical functions page.



Simply touch "Save Chop Points". LD3 will now save your edits, and return to the graphical edit page.

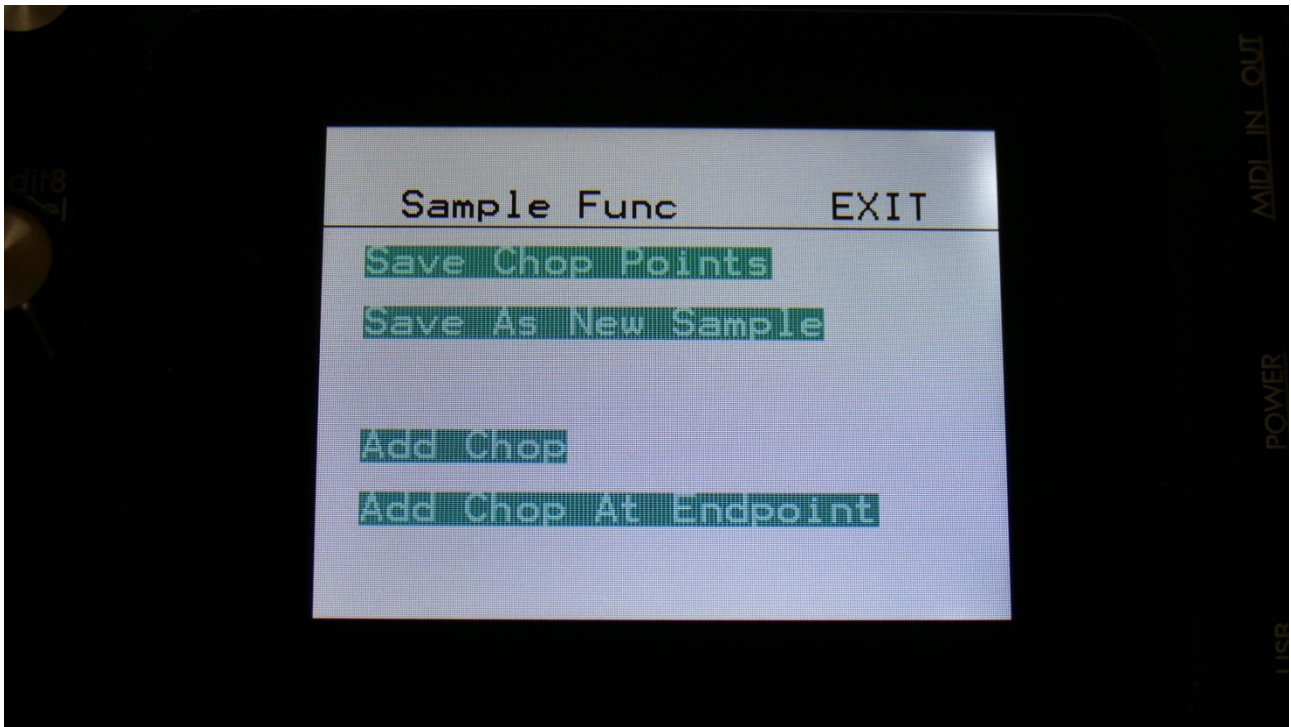
Creating a new sampling from a portion of the selected sampling

If you want to completely remove unused start and/or end portions of a sampling, use a portion of the sampling for a loop, or just wants to copy the whole sampling to another sample bank, you can use this function.

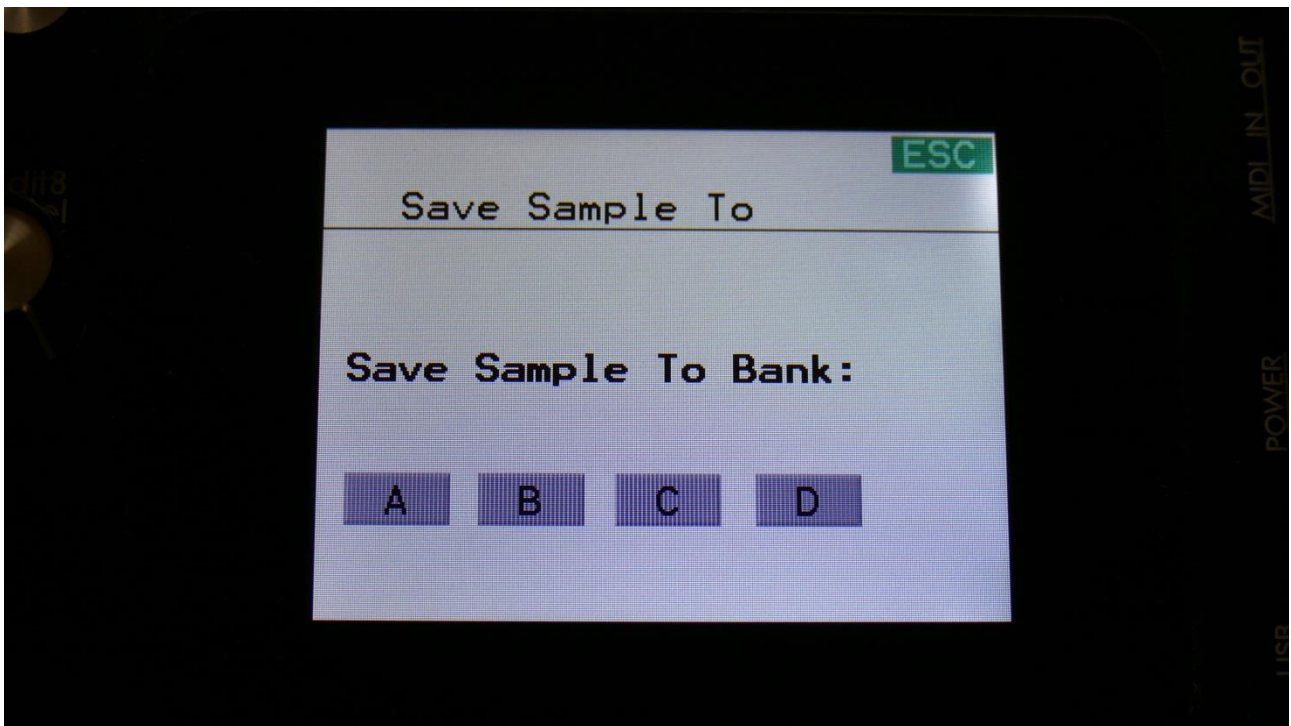


First, set the start and the end points, of the portion of the sampling, that you would like to copy, OR select a chop, if this is what you would like to copy.

Go to the functions page.



Touch “Save As New Sample”. A new page will now open. Select the sample bank, that you would like to copy the sampling/portion of the sampling to, by touching the A, B, C or D touch button.



LD3 will write “Saving Sample...” for some time, and then it will jump back to the functions page.

WaveBuilder

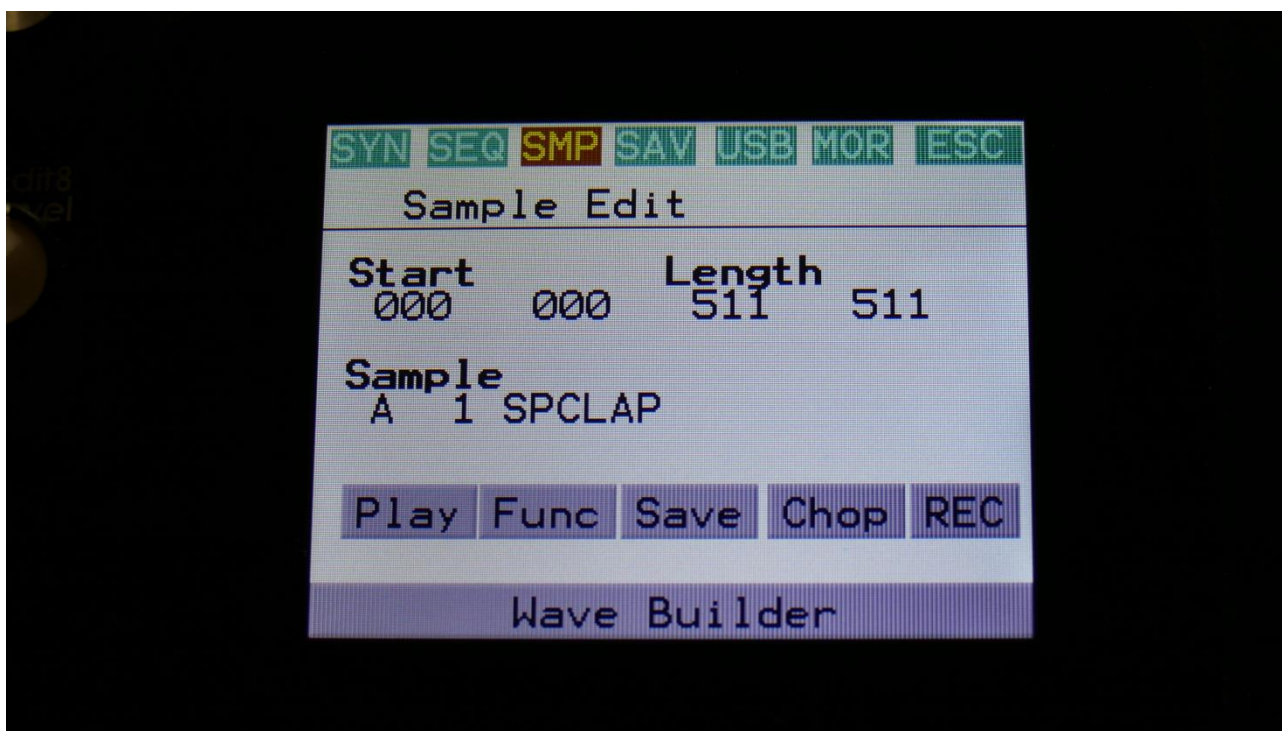
The LD3 WaveBuilder lets you combine up to 8 samplings, to create a wave sequence from these. Whole samplings, parts of the samplings or chop points from the samplings, can be used to build a new wave sequence.

The wave sequence is saved as a new sampling, and a chop point is inserted, at the start point of each of the 8 waves, for easy selection of each partial, using the chop parameter.

WaveBuilder also lets you copy a sampling from one sample bank to another. Just select the sampling in WaveBuilder, set start to 0 and length to 511, and save it to the new bank.

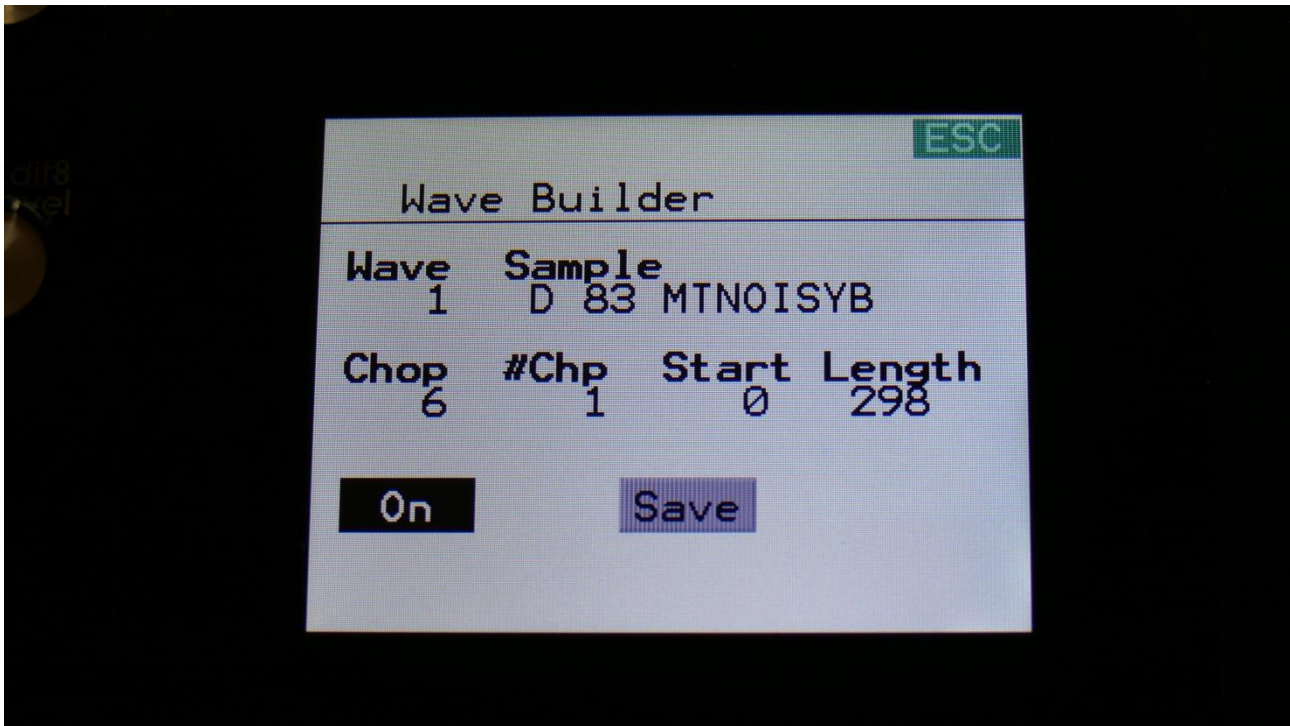
The WaveBuilder always outputs to audio bus 1, so any filters and effects assigned to this bus, will affect the sound.

Enter WaveBuilder:



From the Sample Edit page, touch Wave Builder in the bottom of the page.

You will now enter this page:



Wave: 1 to 8. Selects wave 1 to 8 of the wave sequence to be generated.

Sample: A0 to D2048. Selects the sampling for the selected wave. Edit knob 2 selects the sample bank (A to D), edit knob 3 selects the 100's in the sample number, and edit knob 4 select the 10's and the 1's in the number.

Chop: 1 to 64. Selects the chop point of the selected sampling, to be included in the wave sequence, if the selected sampling holds any chops.

#Chp: 1 to 64. Selects the number of chops to play back, from the selected chop.

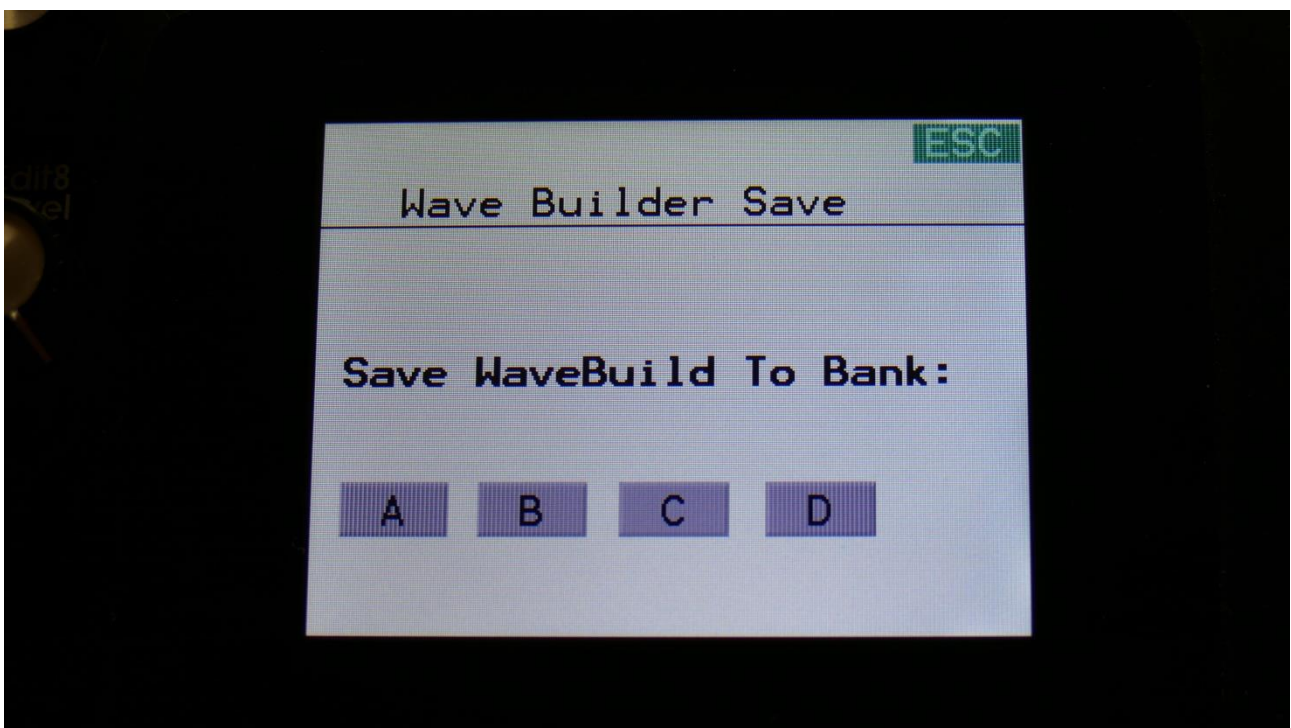
Start: 0 to 511. Sets the start point of the sampling, of the range, that should be included in the wave sequence. If the selected sampling holds chop points, the start point will be inside the selected chop range.

Length: 0 to 511. Sets the length of the sample range, that should be included in the wave sequence. If the selected sampling holds chop points, the maximum length will be inside the selected chop range.

On/Off touch button: Each of the 8 waves can be separately switched on or off. As soon as at least one wave is on, WaveBuilder will start playing back the sequence of the waves. Any waves that are switched off, or has their length set to zero, will be skipped from the wave sequence.

Save touch button: As soon as you have selected samplings and chops, and adjusted the start and lengths, and you are satisfied with the result, you should touch this button, to save the new wave sequence.

This will get you to this page:



Now touch A, B, C or D, to save the wave sequence to the corresponding sample bank.

USB

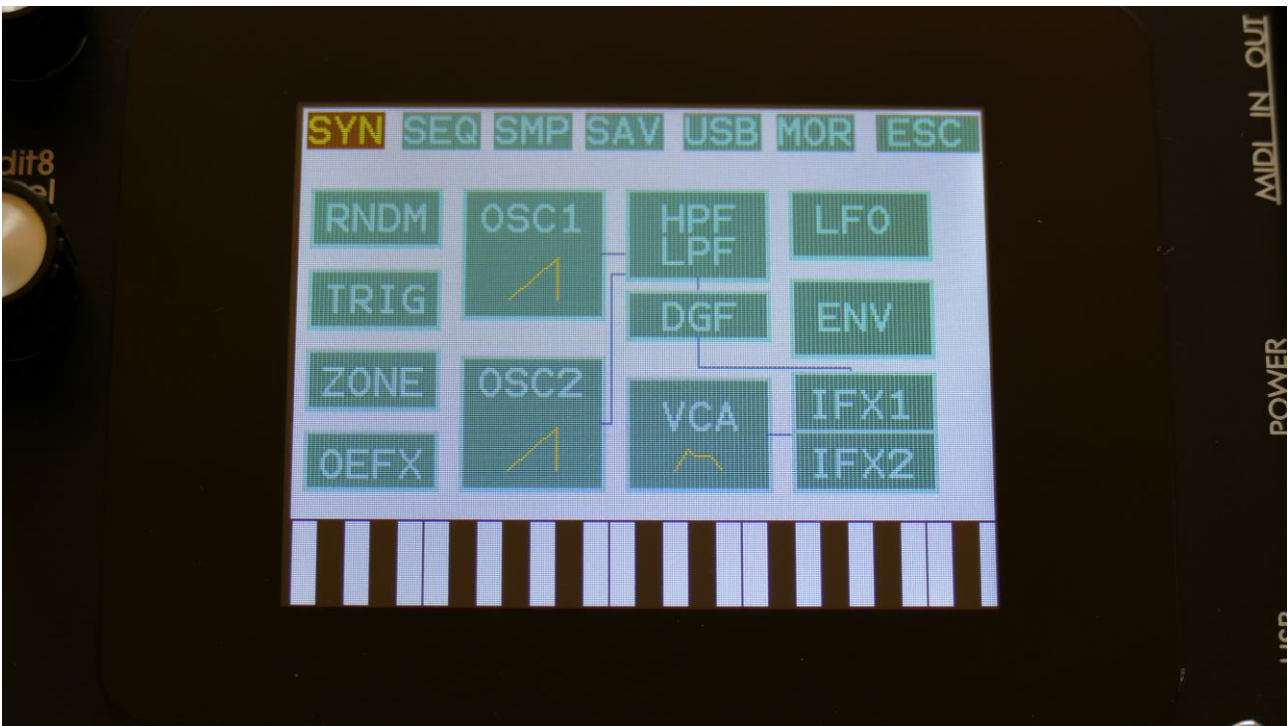
The USB pages lets you access the files and directories of a USB drive, attached to LD3's USB connector.

You can import and export samples as .wav files, import and export presets and songs, update the Id3 firmware, and make new directories.

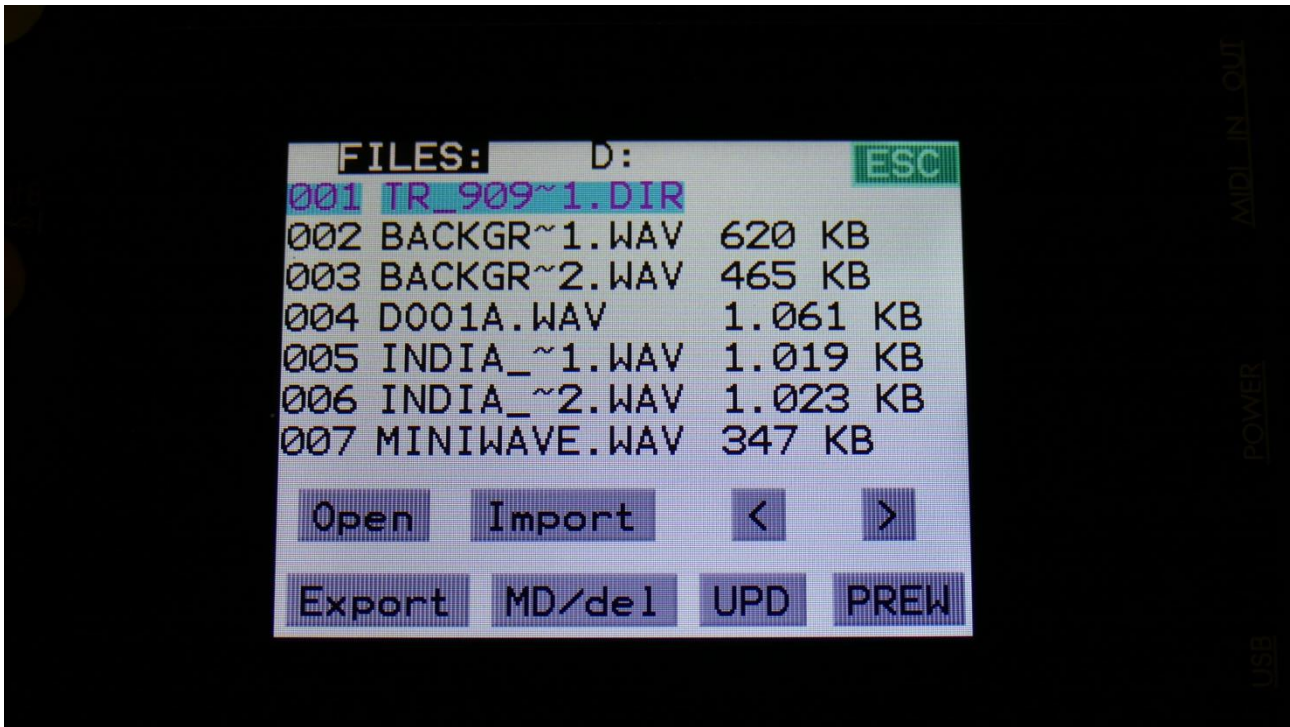
USB drives used with Id3 should be:

- Maximum 32 GB
- FAT formatted

From the Preset Select screen, touch the EDIT field, to enter the Main Synth page:



Touch USB on the navigation bar at the top of the screen.



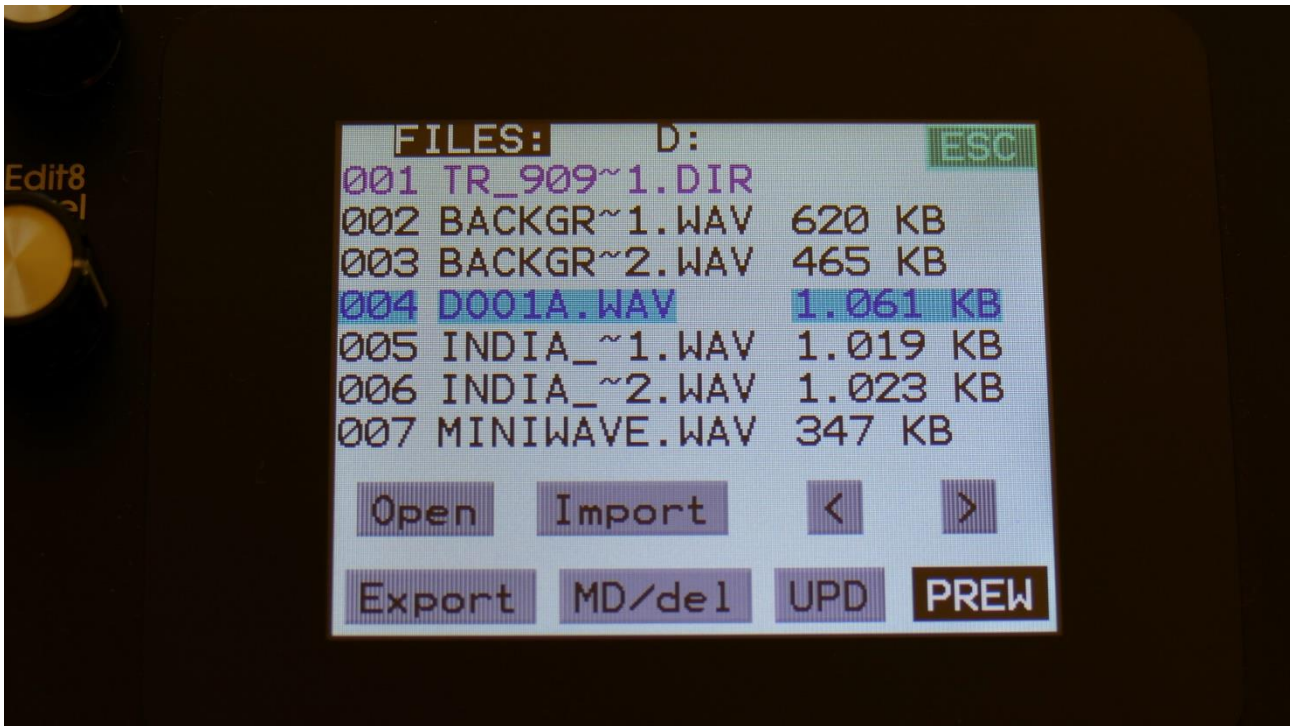
On this first page that you will enter, the root directory of the connected drive is shown. You can see that it is the root directory, by the "D:" in the top of the page. If you have accessed a directory, the name of this will replace "D:".

If you do not see the root directory of the drive, make sure that it is connected properly, and that it has the right specs, as described in the start of this section.

The names you see on the screen, are the names of the files and directories on the drive. File names are in black, and directory names are in purple. The currently selected file has a blue box around it. To select a file, simply touch it.

LD3 will show 7 files/directories at a time. By pushing the arrows, you can select the previous/next 7 files.

USB Sample Preview



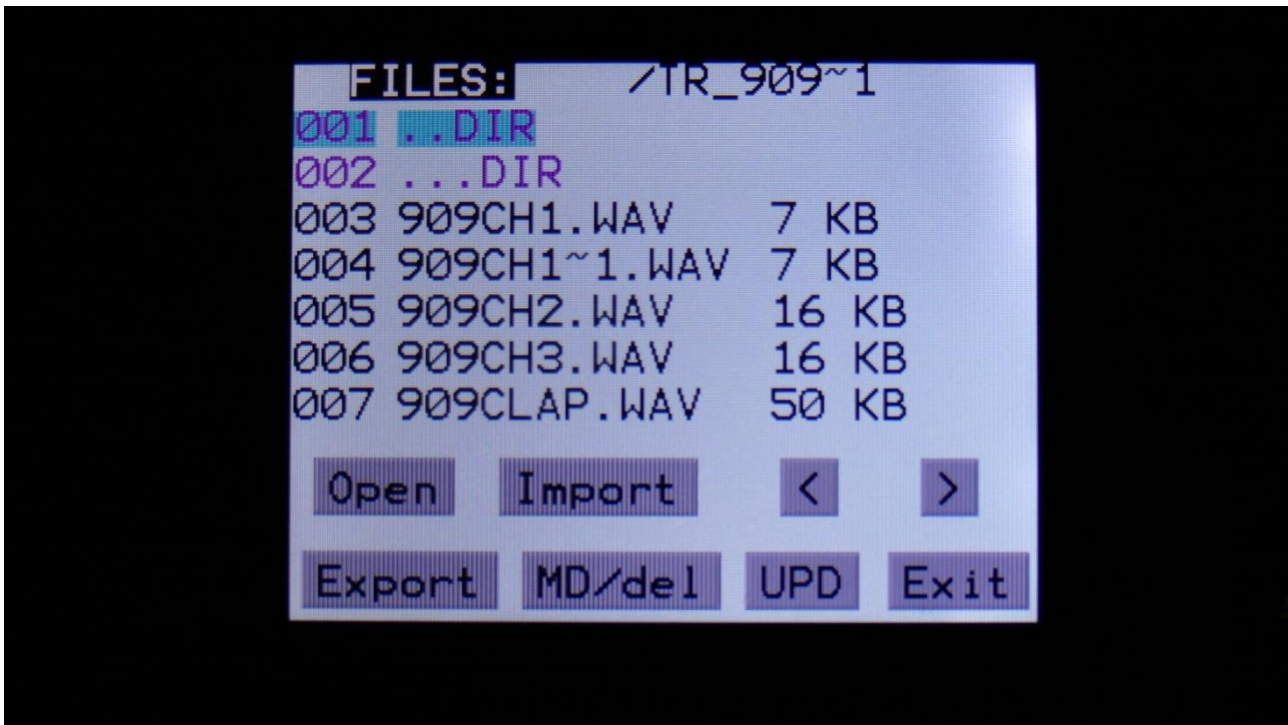
Wav samples can be previewed directly from a USB drive.

Select the sampling, that you would like to preview, by touching the sample name, and touch the PREW touch button. This will first show LOAD, then PREW again, while playing back the first 6 seconds of the sampling.

Open a directory (or folder)

To open a directory, to see what's inside, or to import single files or sub-directories, simply select the directory you would like to open, by touching it, and then touch the "Open" button.

LD3 will now open the directory, and show the first 7 files in it. The "D:" in the top of the screen, will now be replaced by the opened directory's name.



Importing files

Files that can be imported to LD3 are:

- Samplings with the ending .WAV, including LD3 generated chop points, and cue points generated by other gear/computer programs.
- Little deFormer samplings with the ending .LDS, including chop points.
- Presets with the ending .XDP
- Songs with the ending .XDO

Files that are shown, but can't be imported directly are:

- Update files with the ending .HEX

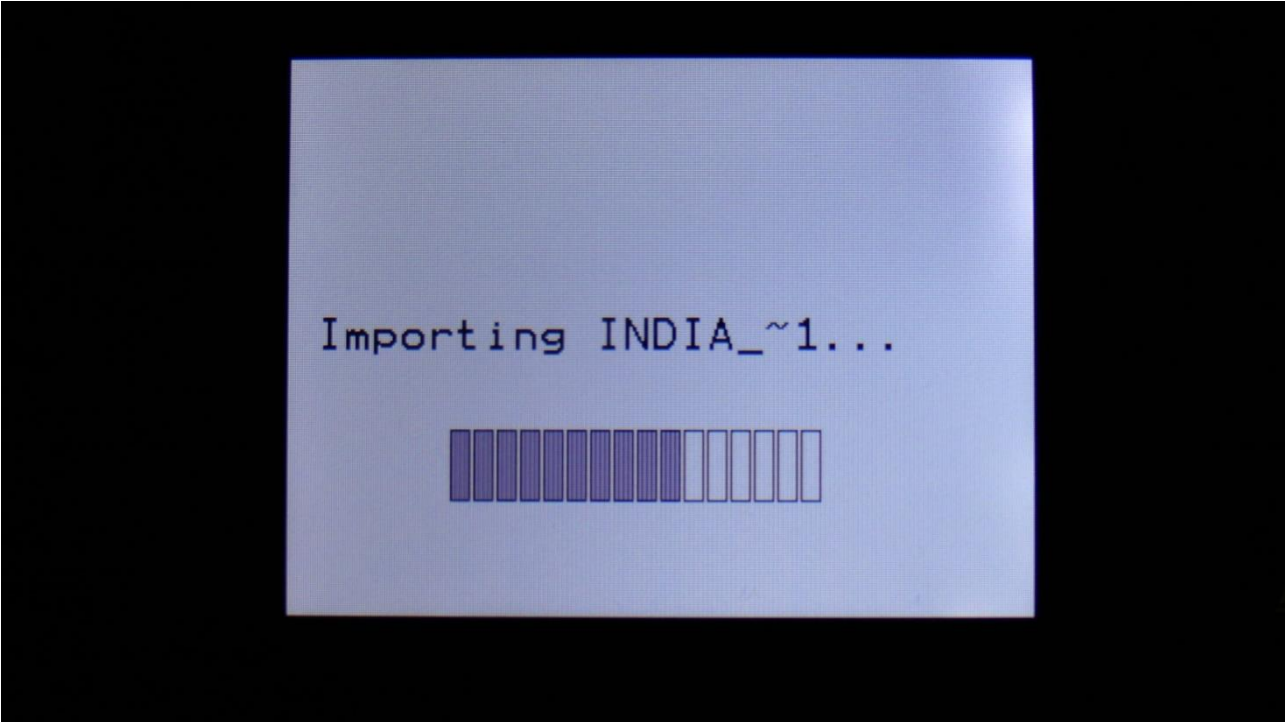
Files with any other ending will not be shown on LD3's screen, regardless of if they are present on the drive.

To import a single sampling or preset/song, select the .WAV/.LDS/.XDP/.XDO file you would like to import, by touching it.

Samples are imported to the sample bank selected on the Sample Edit page.

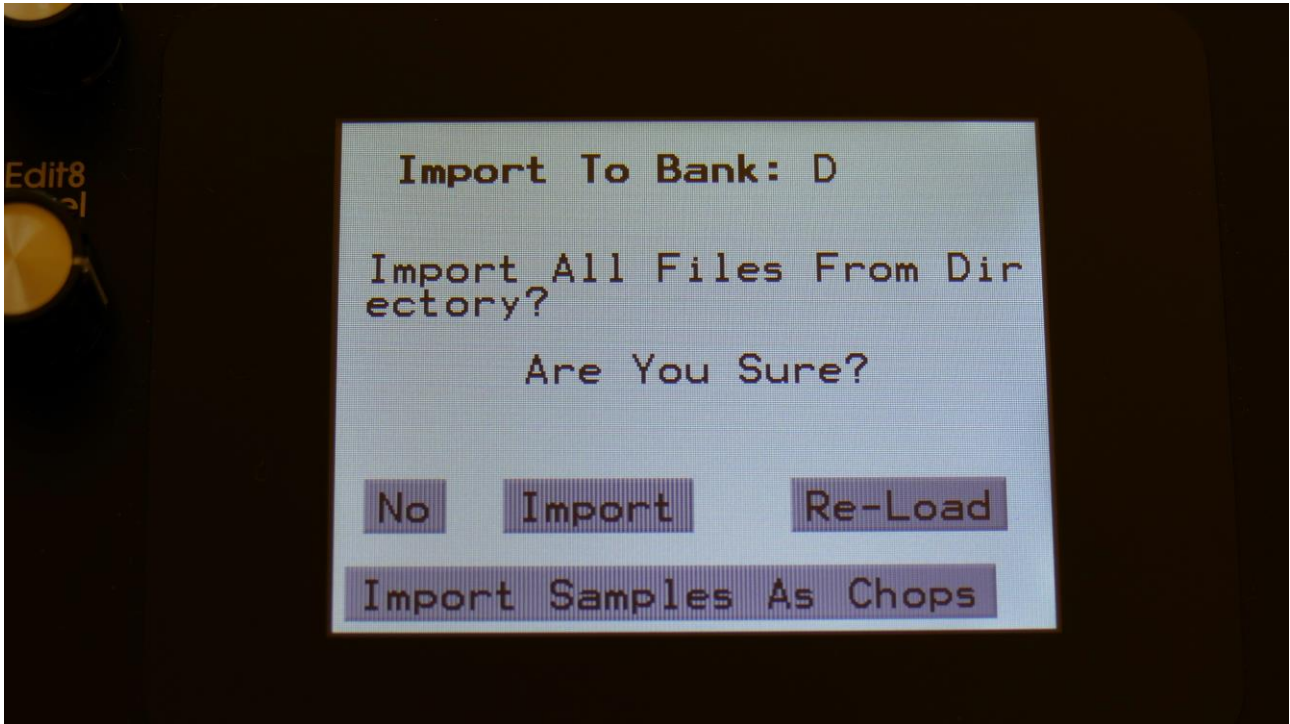
When importing a single preset or song, it is only loaded to the preset/song buffer, and you will have to save it manually, to keep it.

Touch the "Import" button. LD3 will now import the selected file, and show the progress on a progress bar:



Importing multiple files

To import multiple files at one time, select a directory, by touching it. Touch the "Import" button. LD3 will now ask:



Touch "Import" to continue, or "No" to exit.

If you touched Import, LD3 will now import all samples, presets and songs, that the selected directory holds. It will though not import files from any sub-directories.

Samples are imported to the sample bank selected by Edit Knob 1, that is shown in the top of the screen. Presets are imported to the selected preset and forwards, so make sure that you have not selected a preset that you planned to keep, before using this function. Songs are also imported to the selected song and forwards, so make sure that you have not selected a song that you planned to keep, before using this function.

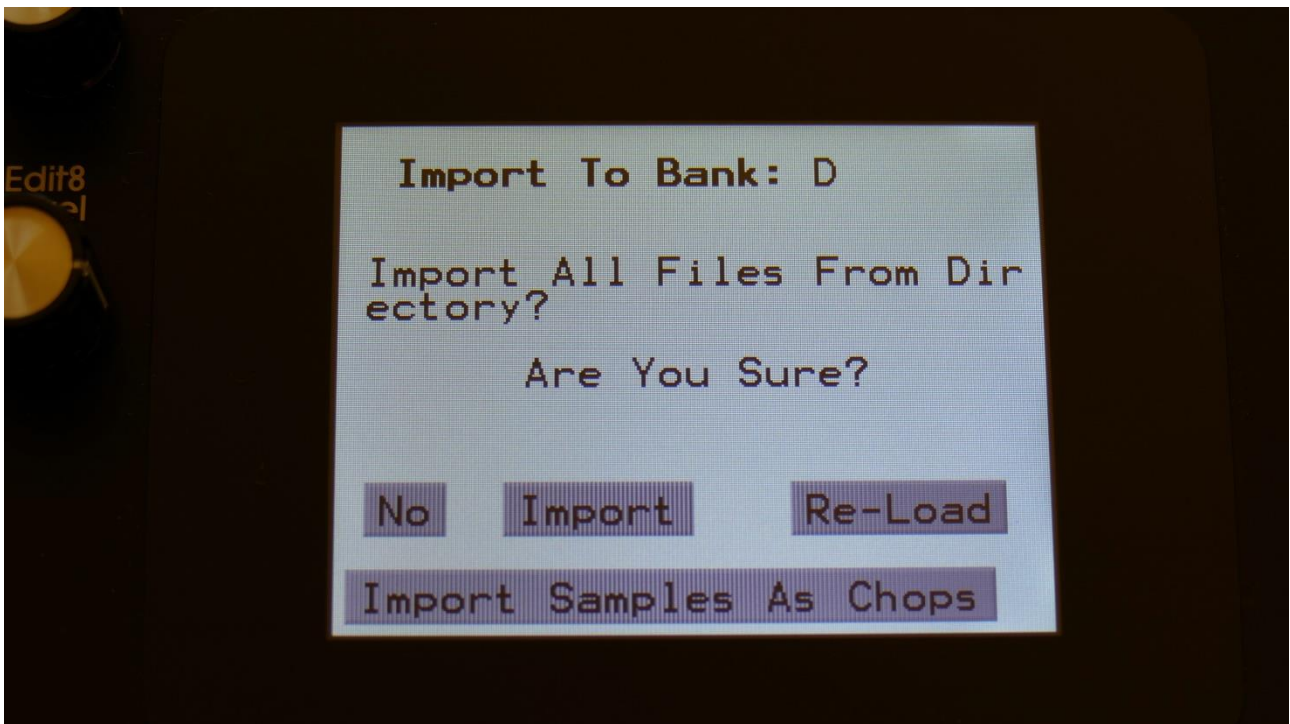
Reload multiple files

A reload function is also available. This will only import presets, songs and samples that has been exported from LD3, or that has been renamed to to fit the LD3 export standard –samples must be named: A0001.WAV, A0002.WAV.....A0220.WAV, B0001.WAV, B0002.WAV.....B0256.WAV, presets and songs must be named: A01.XDP, A02.XDP....A64.XDP, B01.XDP, B02.XDP.....P64.XDP .

This function will import presets and songs to the same locations as they were originally stored to.

Reload will also import samples to the exact same location, as they were placed when exporting. This will make sure, that samples used in your presets, will always be placed in the right locations. Sample locations that are already occupied, will be skipped by Reload, so make sure to erase sample bank A and B, if you want a complete reload.

To Reload multiple files, select a directory, by touching it. Touch the "Import" button. LD3 will now ask:



Touch "Re-Load" to continue, or "No" to exit.

If you touched Re-Load, LD3 will now import all samples, presets and songs, that the selected directory holds, and that are correctly named. It will though not import files from any sub-directories.

Reload does also have other functions:

-On LD3 itself, it is only possible to erase the last recorded sampling of bank A, B, C and D. By exporting all your samplings to a USB drive, and then delete sample bank A, B, C and D on LD3, delete the sample wav files you want to erase on the USB drive, from LD3 itself or on a computer, and reload all the samples. Samples that are erased, will just be left blank.

-If you would like to import a number of wav files in a specific order, or in a specific sample bank, you can rename the samples on a computer to: A0001.WAV, A0002.WAV.....A0220.WAV, B0001.WAV, B0002.WAV.....B0256.WAV, and then import them using Reload.

-If you would like to rearrange the order of samples and/or presets, you can rename the files and reload them. Presets are named: A01.XDP, A02.XDP....A64.XDP, B01.XDP, B02.XDP.....P64.XDP .

Import Multiple Samplings From Directory As one sampling with Chop points

This new function will make it easier than ever, to make your LD3 hold more than 8.192 samplings, in an easy way! Without splicing samples on a computer!

On your computer:

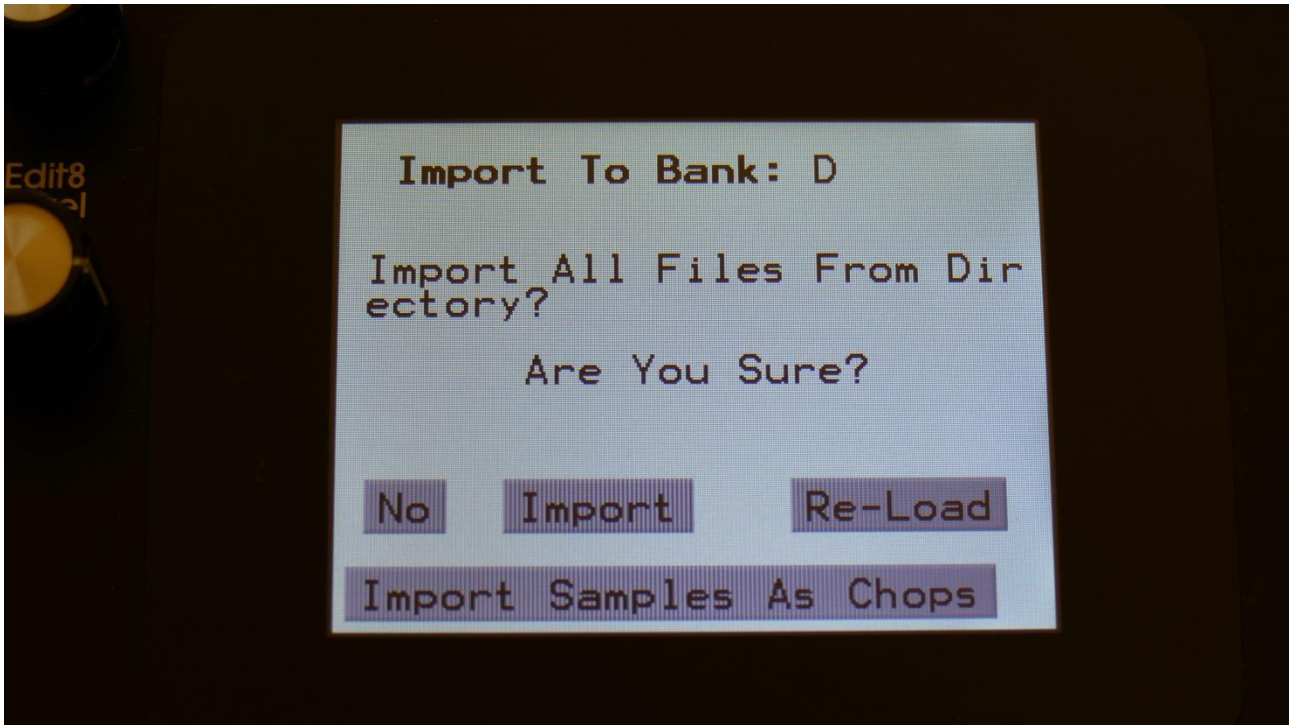
Make a directory (or folder) on an LD3 compatible USB drive. Name the directory with the name, that you want the sampling to have. Copy any single samplings you would like this sampling to contain, into the directory. Max 64 samplings.

Create multiple such directories, if desired.

Unmount the USB drive from your computer, and insert it in the LD3 USB port.

Enter the USB menu, select the directory you just created, and push “Import”.

This screen will now show:



Touch “Import Samples As Chops”.

LD3 will now create one sampling from all of the samplings in the selected directory, and insert a chop point at the start point of each sampling.

Samples are imported to the sample bank selected by Edit Knob 1, that is shown in the top of the screen.

After the import is done, go to the sample edit page, select the sampling you just imported (named from the directory name), and enter the Chop edit page. You will now be able to hear each of the samplings, by selecting chops.

On the synth sampler pages, the procedure is the same. Select the sampling, and select the chop point. Modulate the chop point selection, to create great variation.

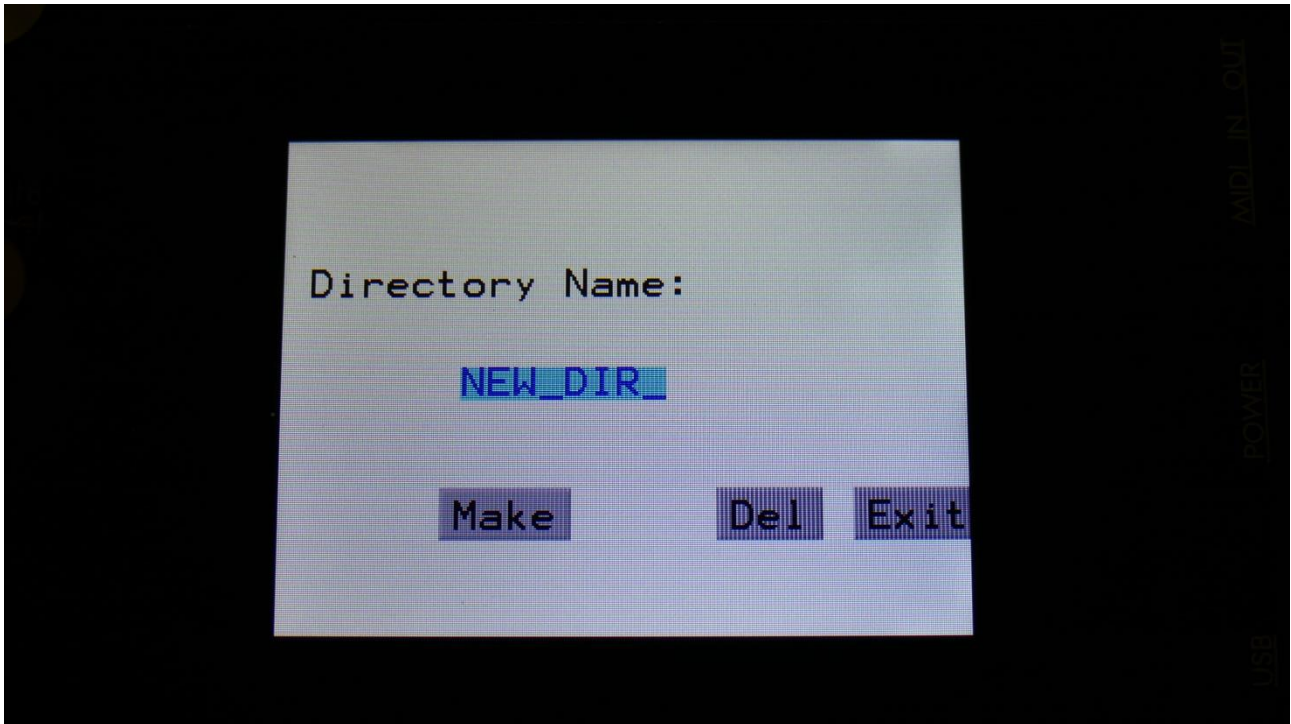
I have found this function especially useful, when working with acoustic drum samplings. There are many acoustic drum sample packs on the net, that has different variations of the same drum sound. Put all the variations of a sound in one directory, import as chops, and modulate the chop

selection with velocity, random or any other modulation source, to make some great dynamic acoustic drum beats.

Make a new directory

A new directory can be created in the root directory, or inside another directory.

To do this, touch the "MD/Del" button.



Turn the 8 edit knobs to select the 8 characters of the name for the new directory. When you are done with this, touch the "Make" button. The new directory will now be created.

You can, of course, also just touch "Exit", if you do not want to make a directory anyway.

Delete file from USB drive

It is possible to delete a file from the attached USB drive, directly from LD3.

To do this, select the file you would like to delete, and touch the "MD/Del" button.



Now touch the "Del" button.

Export samples, presets and songs to a USB drive

The samples, presets and songs held in LD3's FLASH memory can be exported to a USB drive, for back-up, or for use with other gear or computers. LD3 exports samples as standard 16 bit, 44.1 KHz, native PCM .wav files, so they can be used with any other gear, that supports this format. LD3 exports presets and songs in its own .xdp/.xdo format, so these can at writing moment only be used by LD3.

The factory sample bank can't be exported.

To enter the export page, touch the "Export" button:



All exports are done to the currently open directory. There are 4 different possibilities for exporting:

"1Prs"

Export 1 preset (song). Touching this button will export the currently selected preset. If LD3 are in song mode, it will export the selected song.

"AlPr"

Export All Presets/Songs . Touching this button will export all programmed presets. If LD3 are in song mode, it will export all programmed songs.

"1Smp"

Export 1 Sample. Touching this button will export the selected sample. Turn Edit Knob 1 to select the sample bank of the sample to be exported, and turn Edit Knob 2 to select the "100's" of the sample number, and Edit Knob 3 to select the 10's and 1's.

If you have selected any of the factory samplings, this will do nothing.

"AlSm"

Export All Samples. Touching this button will export all samples of the selected sample bank. The Sample Bank can be selected, by turning Edit Knob 1.

If the selected sample bank contains any factory samplings, these will not be exported.

Updating the LD3 firmware

Now and then updates will be available for LD3, that adds new functionality and fixes bugs. These will always be available for download at:

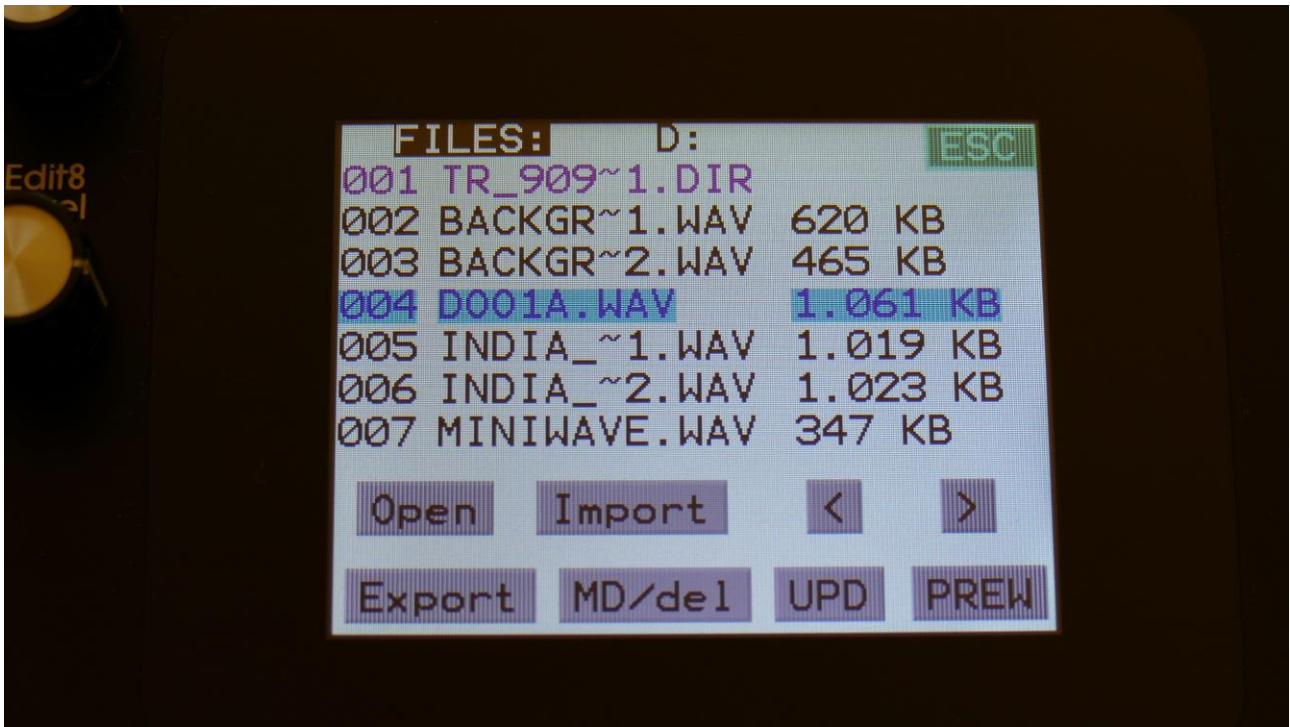
<http://www.gotharman.dk>

PLEASE NOTICE: The Spazeboard6 firmware versions will always have a number, that starts with 8, like: LD3_8093.HEX

To update LD3, you must have a computer with an internet connection, and an LD3 compatible USB drive ready. See the start of this section, for which USB drives that are LD3 compatible.

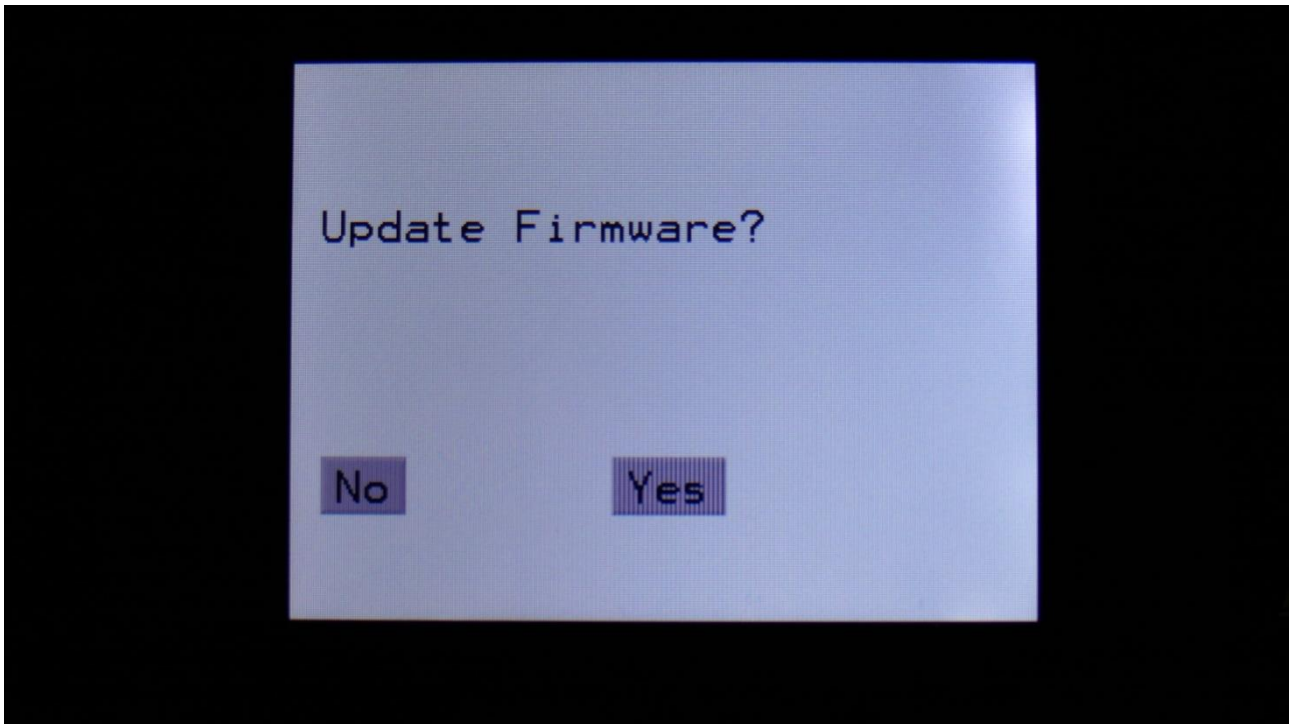
Then you should follow these steps:

1. Download the update file of the latest update, from the LD3 Updates site, to your computer.
2. Connect the USB drive to your computer.
3. Make sure that the USB drive is FAT formatted.
4. Create a directory in the root directory of this USB drive, that is named "UPDATE".
If the USB drive already contains a directory called UPDATE, please delete all files inside this.
5. Copy the update file from your computer to the USB drive UPDATE directory. Make sure that no other files are present in this folder, and please don't rename the files.
6. Eject and remove the USB drive from your computer, and connect it to LD3 's USB connector.
7. Turn LD3 on, if it isn't already turned on.
8. From the preset select screen, touch the EDIT field.
9. Touch USB.
10. Wait for LD3 to register the USB drive, so all files and folders are presented on the screen.



11. Touch the "UPD" button.

12. LD3 will now search for the "UPDATE" folder and a valid update file. If it finds both, it will ask: "Update Firmware?"

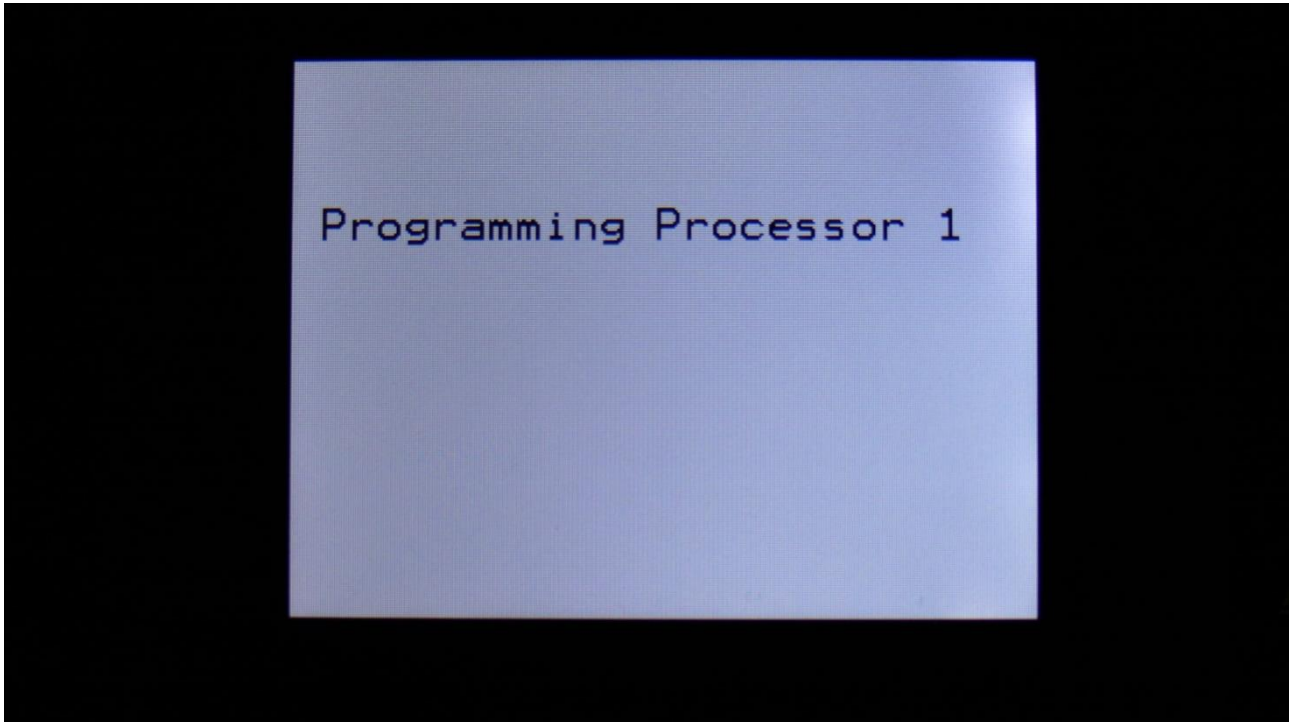


13. Touch "Yes" to proceed with updating, or "No" if you regret.

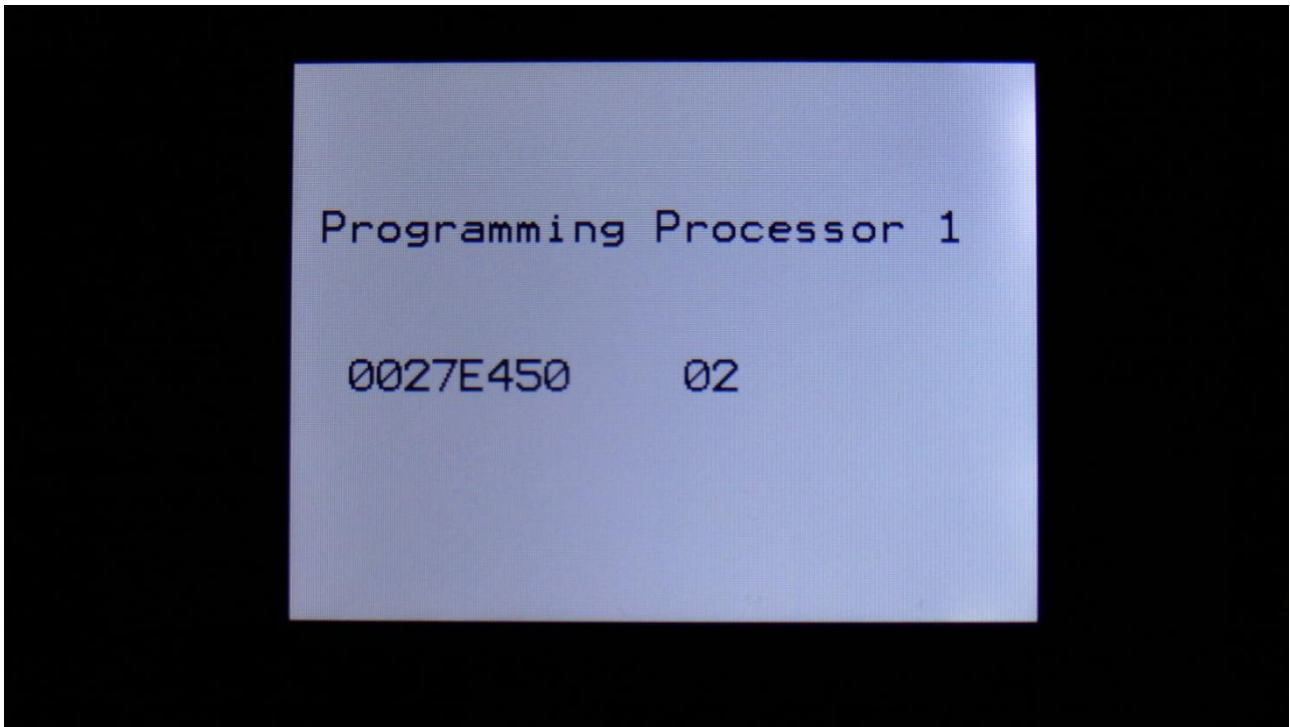
14. If you pushed Yes in step 13, LD3 will now start to get the update file, and program its processor.

When it starts to program the processor, it will look like it halts for a couple of minutes. This is because it does halt, when erasing the program memory of a processor.

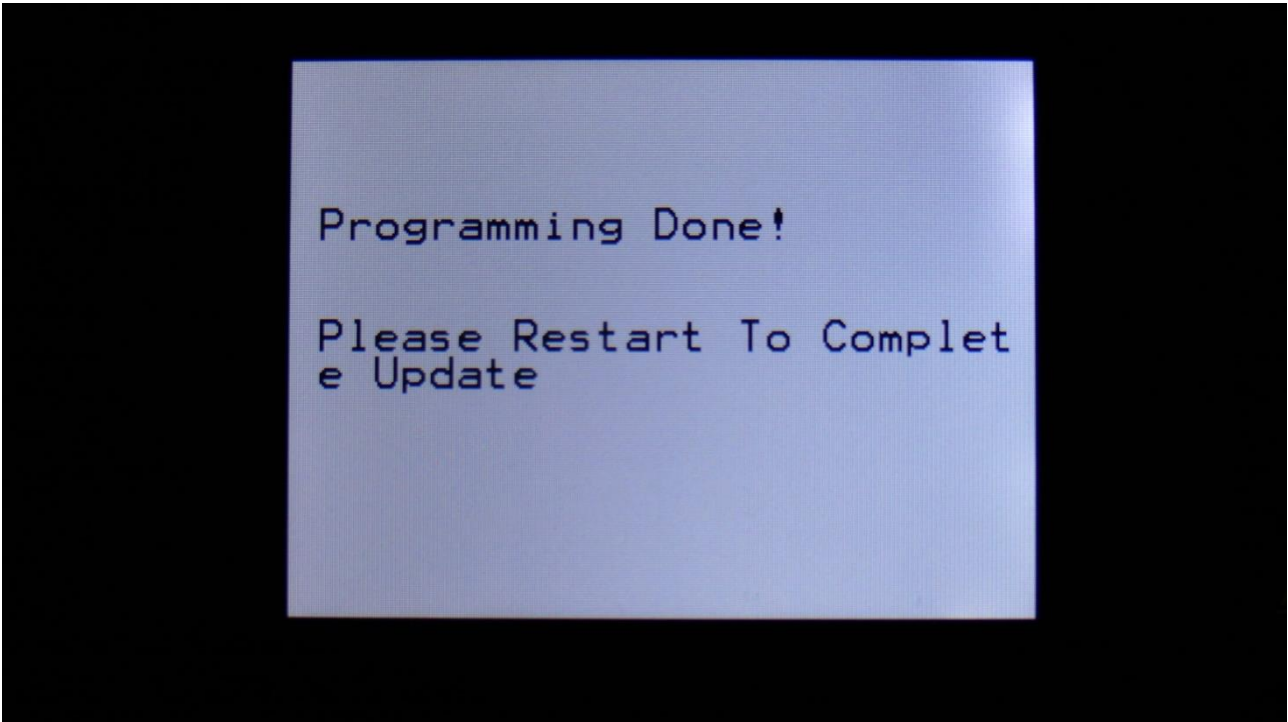
Erasing the program memory of the processor:



Programming processor (counter in the bottom counts up):

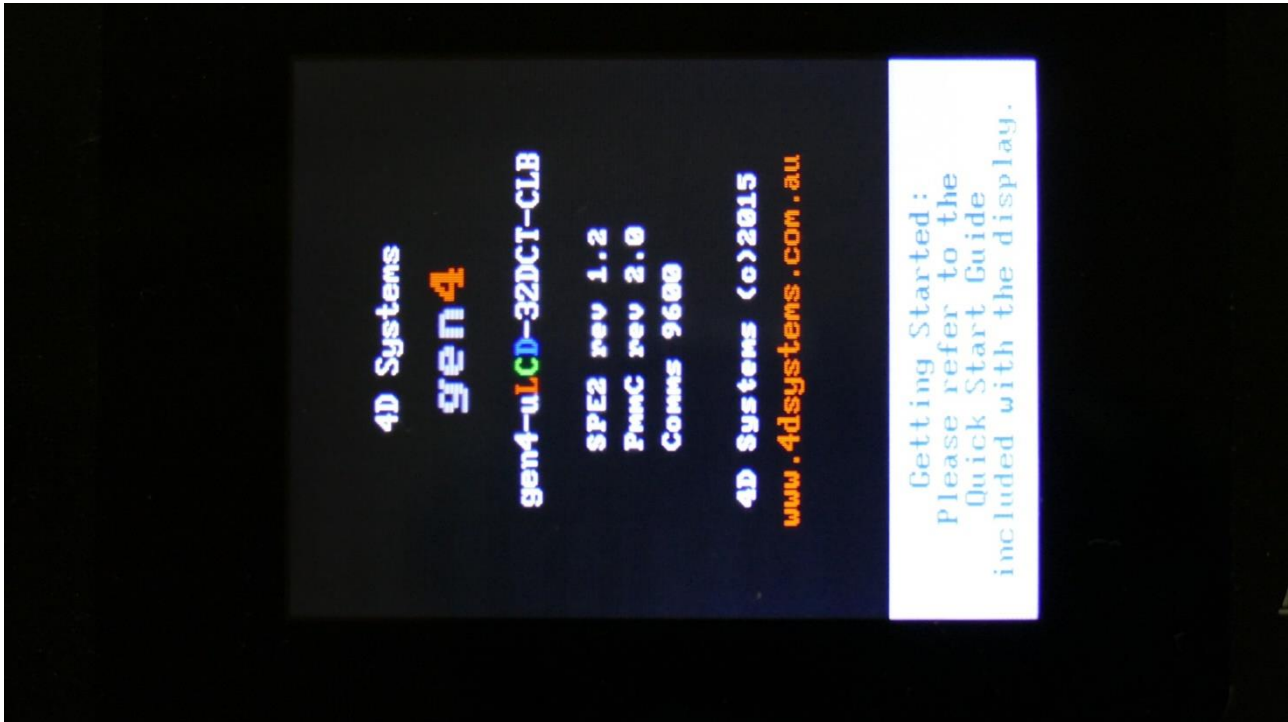


15. When programming is done, LD3 will ask you to restart it. Please turn LD3 off and then on again.



16. When you turn on LD3 after an update, it will need to finalize this. It will show this by turning 10 step button LED's on and off sequentially, one by one. After it has done this for a little while, it will start normally and are ready to use.

While it is finalizing the update, an initial text will show on the display:



If something goes wrong:

-If the power should go off, while LD3 is getting the update from the USB drive, and programming the processors, simply start over again from step 7.

-If the power should go off, while LD3 is finalizing the update (3 LED's turn on and off), simply turn LD3 on again, and it will now finalize the update.

MIDI Specs

Receives:

-Note on/off's on the LD3 global MIDI channel and on the individual part MIDI channels in Multi-timbral mode. These can trigger the parts and be recorded by the sequencer, when it is in realtime rec mode.

-MIDI CC 1, 2, 4, 5, 7, 8, 9, 10, 11 and 12 on the LD3 global MIDI channel and on the individual part MIDI channels in Multi-timbral mode. All parameters that are set up to be controlled by Edit Knob 1 to 8, are also controlled by these MIDI CC's.

Morph Knob: CC1

Seq Morph Knob: CC2

Edit Knob 1: CC4

Edit Knob 2: CC5

Edit Knob 3: CC8

Edit Knob 4: CC9

Edit Knob 5: CC10

Edit Knob 6: CC11

Edit Knob 7: CC12

Edit Knob 8: CC7

CC16, touch screen keyboard Y-position on the part zone midi channel, regardless of, if LD3 are in multi-timbral mode or not.

MIDI CC's 17 to 61 as modulation sources.

-Pitch Bend messages on the LD3 global MIDI channel and on the individual part MIDI channels in Multi-timbral mode. Controls the oscillators and samplers of part 1 to 16, and any parameter that has pbend set as its modulation source.

-MIDI clock and start/stop messages are received if the sequencer are in external sync mode.

-Program change and bank change (MIDI CC 32), if Prgr is switched on at the Common page.

Accepted bank changes:

0: Preset bank A and B.

1: Preset bank C and D.

2: Preset bank E and F.

- 3: Preset bank G and H.
- 4: Preset bank I and J.
- 5: Preset bank K and L.
- 6: Preset bank M and N.
- 7: Preset bank O and P.

Any other bank change messages will be ignored

Transmits:

- MIDI CC 1, 2, 4, 5, 7, 8, 9, 10, 11 and 12 on the LD3 global MIDI channel, or on the part MIDI channel, from the knobs on the Preset/Song Select screen, and on any other pages, that has no other Edit Knob functions.

CC16, touch screen keyboard Y-position on the global midi channel.

-Note on/off's on any MIDI channels from the sequencer/part system.

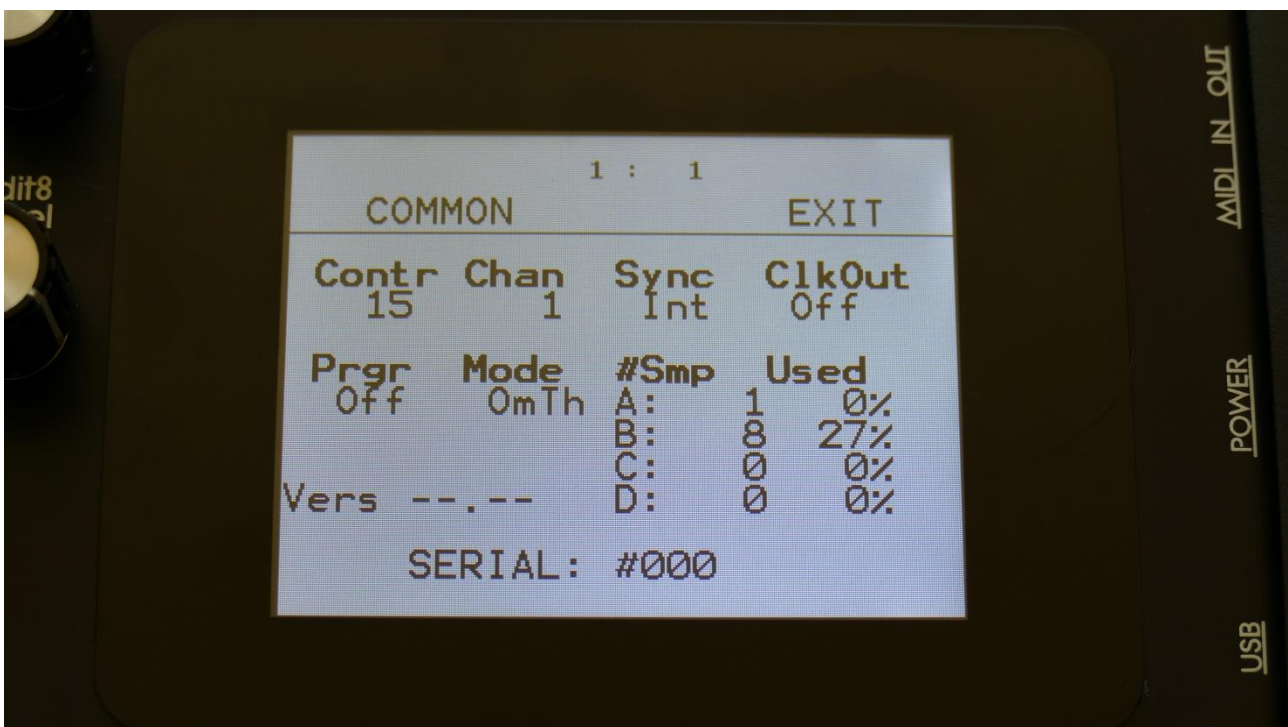
-MIDI CC's on any MIDI channel from the sequencer controller tracks.

Parameters CC control

Many of the Spazeboard6 parameters can be hard-assigned to MIDI CC's, for direct control from an external MIDI controller. The same parameters can also be controlled directly from the sequencer controller tracks, and movements of these can be realtime recorded to the controller tracks.

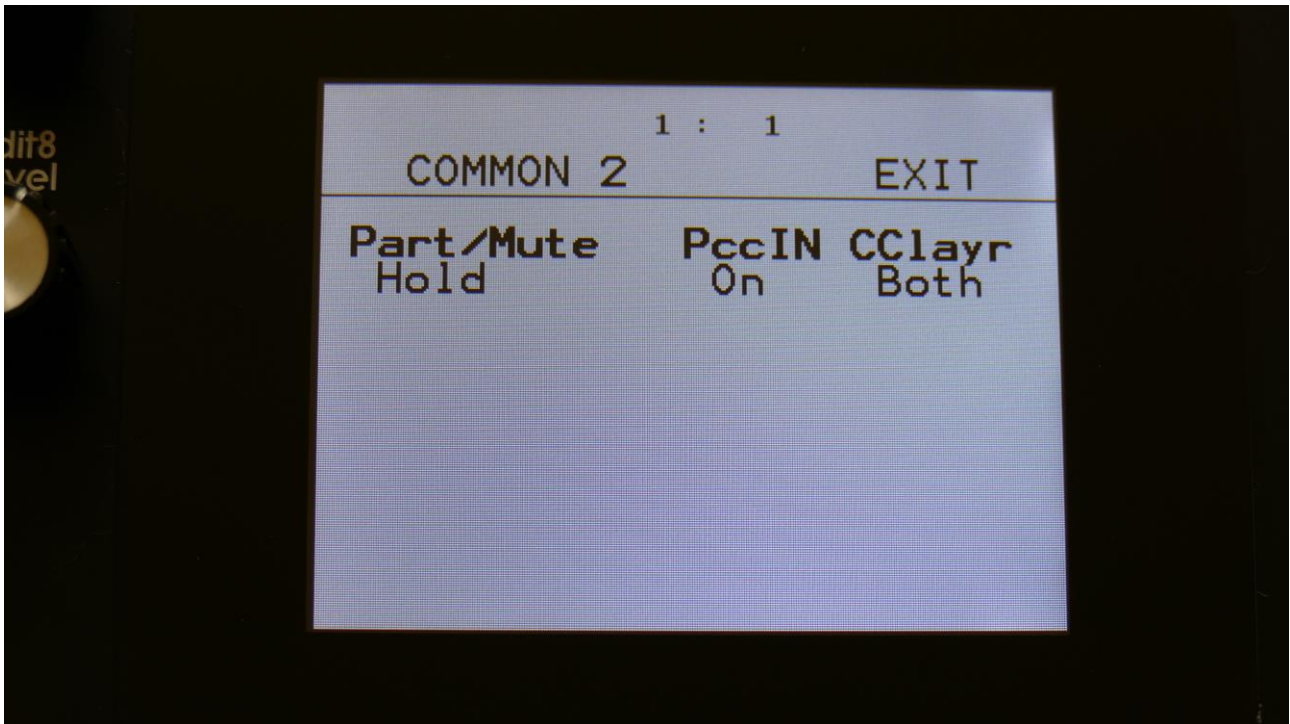
For part parameters, MIDI channel are equal to part number.

To activate the CC control, first enter the MOR>COMMON page:



Here you must set the Mode parameter (MIDI In Mode) to either Omni, OmTh (Omni Thru), Mult (Multi-Timbral) or MuTh (Multi-Timbral/Thru). The CC control will not work properly in Sel (Selected) mode.

Then exit from the COMMON page, and enter the COM2 page:



On this page, switch the **PccIN** parameter (Parameter CC Input) to On.

Set the **CClayr** parameter to Sel, if you would only like to control the selected Morph layer, or to Both, if you would like to control both Morph layers A and B at the same time.

Now you will be able to control many of the Spazeboard6 parameters from an external MIDI device, and from the Sequencer Controller Tracks.

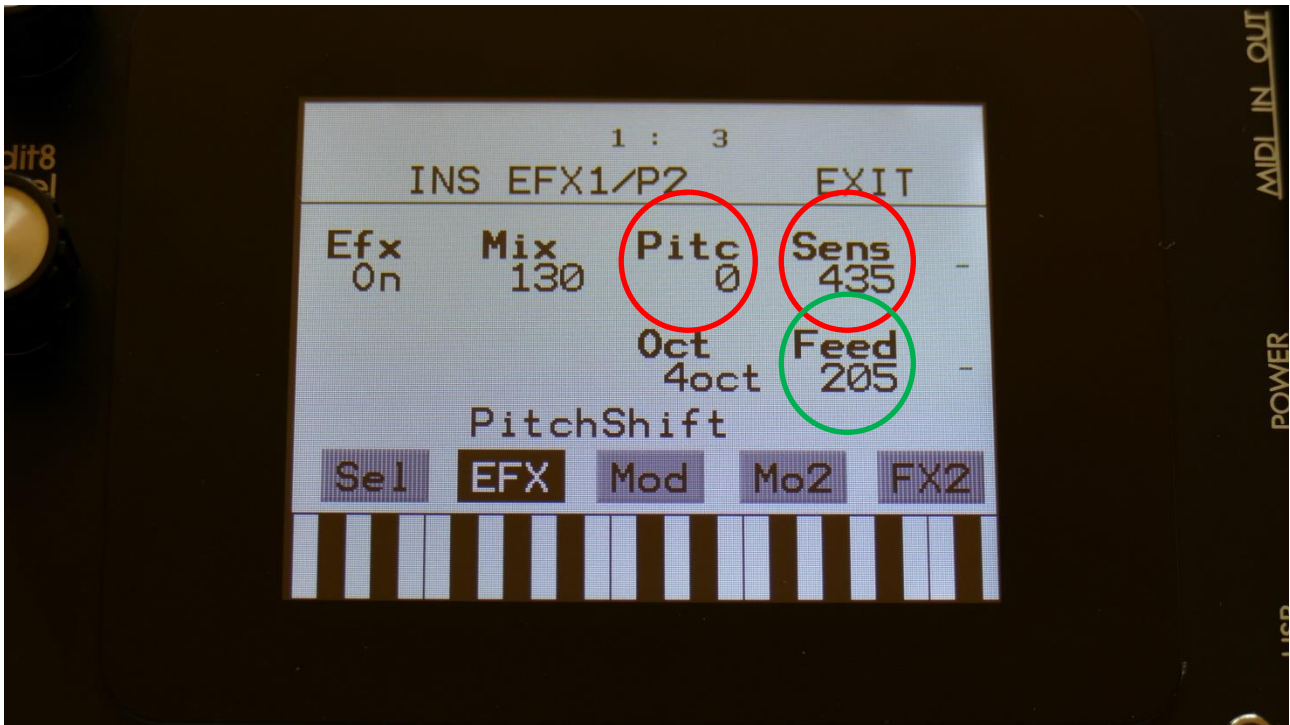
On the following pages, you will find a list of the controllable parameters.

PARAMETER	CC#	PART NUMBER
Oscillator1/Sampler Tune	17	MIDI channel 1-8
Oscillator1/Sampler Fine Tune	18	MIDI channel 1-8
Oscillator1 Wave/Sampler Start	19	MIDI channel 1-8
Oscillator1 PW/Sampler Length	20	MIDI channel 1-8
Oscillator1/Sampler FM	21	MIDI channel 1-8
Oscillator1/Sampler Porta	22	MIDI channel 1-8
Sampler Chop Select	23	MIDI channel 1-8
Oscillator1/Sampler Pitch Mod	24	MIDI channel 1-8
Oscillator1 WaveMod/Sampler ChopMod	25	MIDI channel 1-8
Oscillator1 PWM/Sampler StartMod	26	MIDI channel 1-8
Oscillator1/Sampler FM Mod	27	
Oscillator 2 Tune	28	MIDI channel 1-6
Oscillator 2 Fine Tune	29	MIDI channel 1-6
Oscillator 2 Waveform	30	MIDI channel 1-6
Oscillator 2 PW	31	MIDI channel 1-6
Oscillator 2 PitchMod	33	MIDI channel 1-6
Oscillator 2 PWM	34	MIDI channel 1-6
VCF HPF Cutoff	35	MIDI channel 1-6
VCF HPF Reso	36	MIDI channel 1-6
VCF LPF Cutoff	37	MIDI channel 1-6
VCF LPF Reso	38	MIDI channel 1-6
VCF Osc1 Level	39	MIDI channel 1-6
VCF Osc2 Level	40	MIDI channel 1-6
VCF Ring Modulator Level	41	MIDI channel 1-6
VCF FFM	42	MIDI channel 1-6
VCF G-RAY Feed	43	MIDI channel 1-6
VCF Boost	44	MIDI channel 1-6
VCF HpCutMod1	45	MIDI channel 1-6
VCF HpCutMod2	46	MIDI channel 1-6
VCF LpCutMod1	47	MIDI channel 1-6
VCF LpCutMod2	48	MIDI channel 1-6
VCF HpResoMod	49	MIDI channel 1-6
VCF LpresoMod	50	MIDI channel 1-6
VCF FFM Mod	51	MIDI channel 1-6
VCF G-RAY FeedMod	52	MIDI channel 1-6

PARAMETER	CC#	PART NUMBER
Insert EFX1 Mix	83	MIDI channel 1-6
Insert EFX1 Parameter 1	84	MIDI channel 1-6
Insert EFX1 Parameter 2	85	MIDI channel 1-6
Insert EFX1 Parameter 3	86	MIDI channel 1-6
Insert EFX1 Parameter 1 Mod1	87	MIDI channel 1-6
Insert EFX1 Parameter 1 Mod2	88	MIDI channel 1-6
Insert EFX1 Parameter 2 Mod	89	MIDI channel 1-6
Insert EFX1 Parameter 3 Mod	90	MIDI channel 1-6
Insert EFX2 Mix	91	MIDI channel 1-6
Insert EFX2 Parameter 1	92	MIDI channel 1-6
Insert EFX2 Parameter 2	93	MIDI channel 1-6
Insert EFX2 Parameter 3	94	MIDI channel 1-6
Insert EFX2 Parameter 1 Mod1	95	MIDI channel 1-6
Insert EFX2 Parameter 1 Mod2	96	MIDI channel 1-6
Insert EFX2 Parameter 2 Mod	97	MIDI channel 1-6
Insert EFX2 Parameter 3 Mod	98	MIDI channel 1-6
LFO Rate	99	MIDI channel 1-16
LFO Wave	100	MIDI channel 1-16
LFO RateMod	101	MIDI channel 1-16
LFO WaveMod	102	MIDI channel 1-16
Output EFX1 Mix	103	-
Output EFX1 Parameter 1	104	-
Output EFX1 Parameter 2	105	-
Output EFX1 Parameter 3	106	-
Output EFX1 Parameter 1 Mod1	107	-
Output EFX1 Parameter 1 Mod2	108	-
Output EFX1 Parameter 2 Mod	109	-
Output EFX1 Parameter 3 Mod	110	-
Output EFX2 Mix	111	-
Output EFX2 Parameter 1	112	-
Output EFX2 Parameter 2	113	-
Output EFX2 Parameter 3	114	-
Output EFX2 Parameter 1 Mod1	115	-
Output EFX2 Parameter 1 Mod2	116	-
Output EFX2 Parameter 2 Mod	117	-
Output EFX2 Parameter 3 Mod	118	-

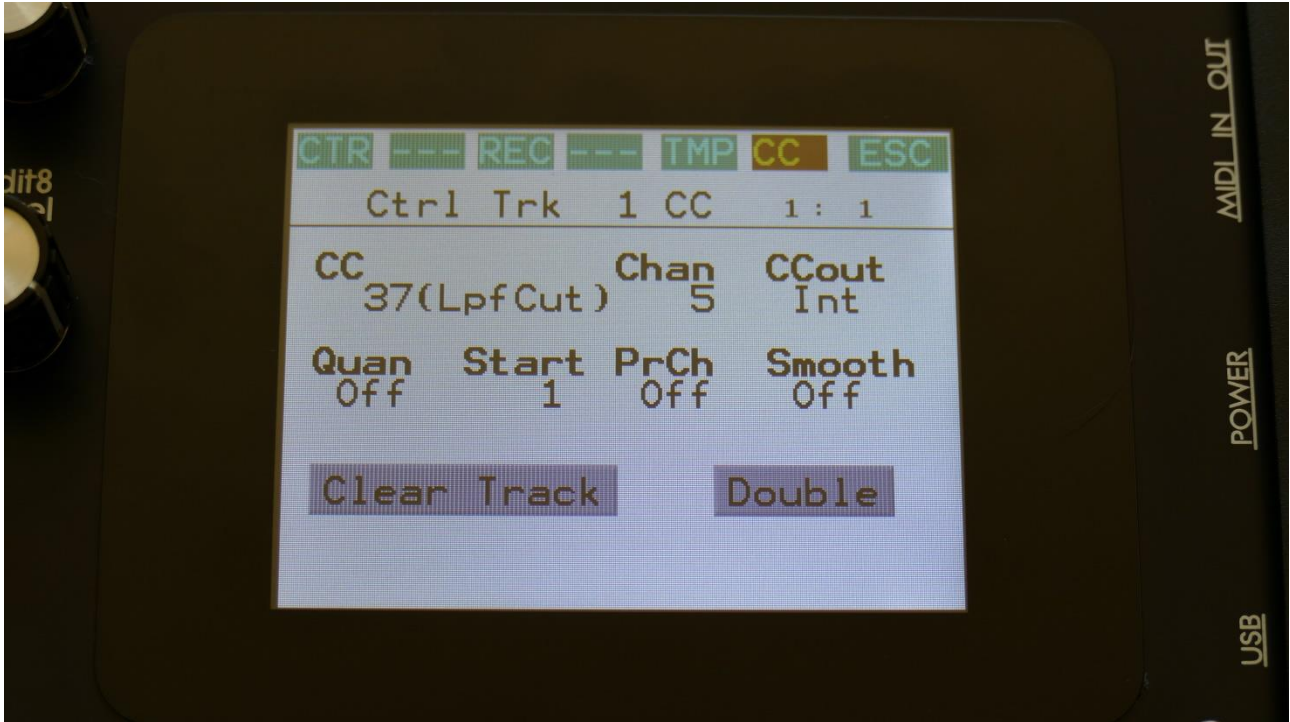
Effects Parameters 1, 2 and 3

The effects parameters 1 and 2, are the 2 parameters located to the right of the Mix parameter, on the effects main pages. Parameter 3 are the lower rightmost parameter:



Controlling the CC parameters from the Sequencer Controller Tracks

Enter a sequencer controller track, and go to the CC sub page:



Set the CC parameter, to the parameter, that you want to control with the controller track. A shorted version of the parameter name, will be written in parenthesis.

Then set the Chan parameter to the part number, that you would like to control, if it is a part related parameter.

At last, set the CCout to internal.

It is **important** to set the parameters up, using this sequence. If you set the CCout parameter to internal first, and then set the CC parameter, all the parameters that you pass, will be affected by the controller track.

Now go to the CTR page, and control the parameter.

Installing SPAZEboard6

This chapter will describe how to install Spazeboard6 into your LD3, in case you didn't purchase it with Spazeboard6 already installed.

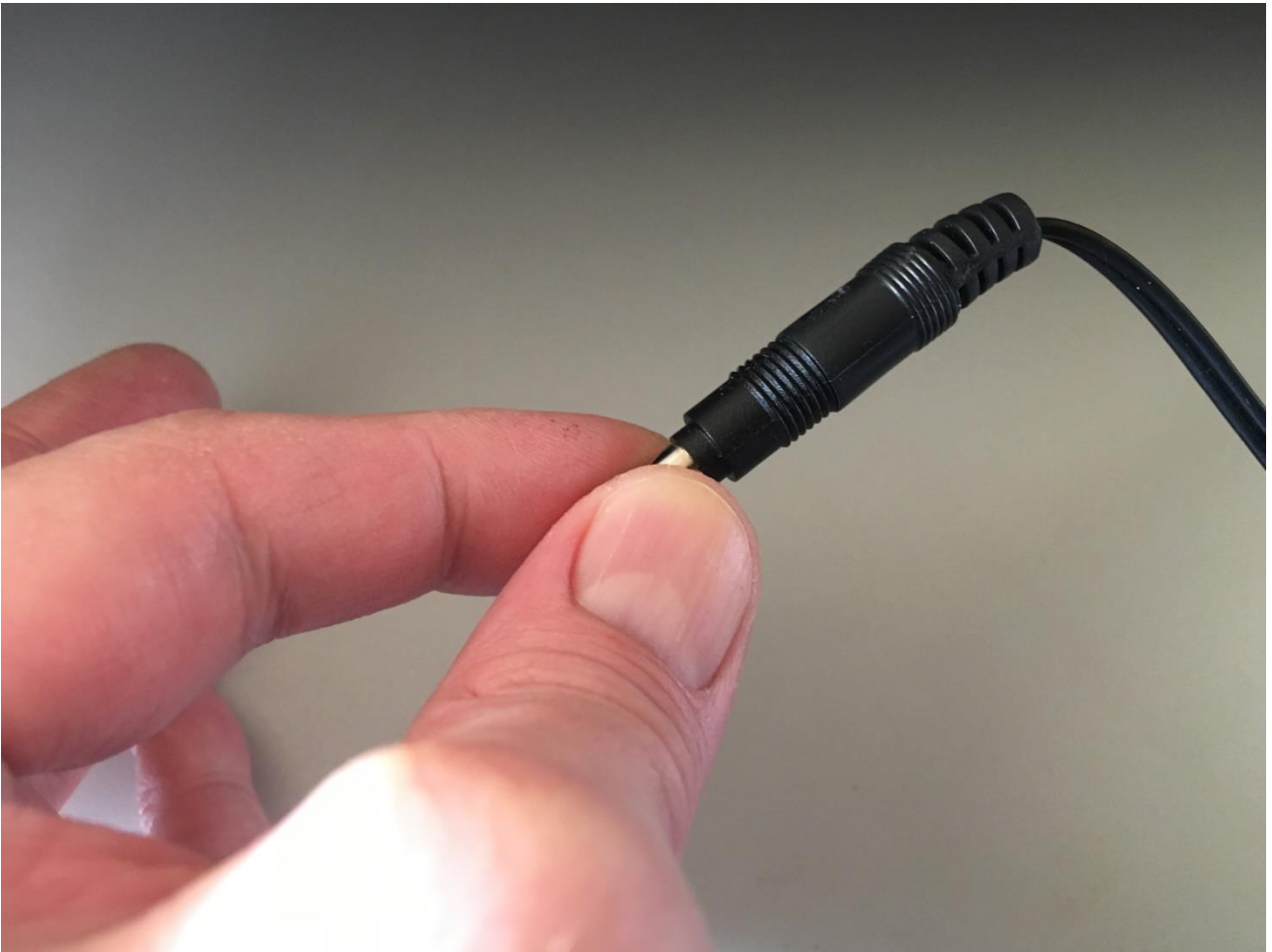
PLEASE NOTICE: In order to install Spazeboard6 in your LD3, you will need to desolder an SMT chip. If you do not feel comfortable doing this, please ship your LD3 to Gotharman's for the installation, or ask your local musical instruments repair shop, if they can do it.

You will need the following tools:

- A Phillips screwdriver
- A 2.5mm Allen screwdriver
- De-soldering tools.



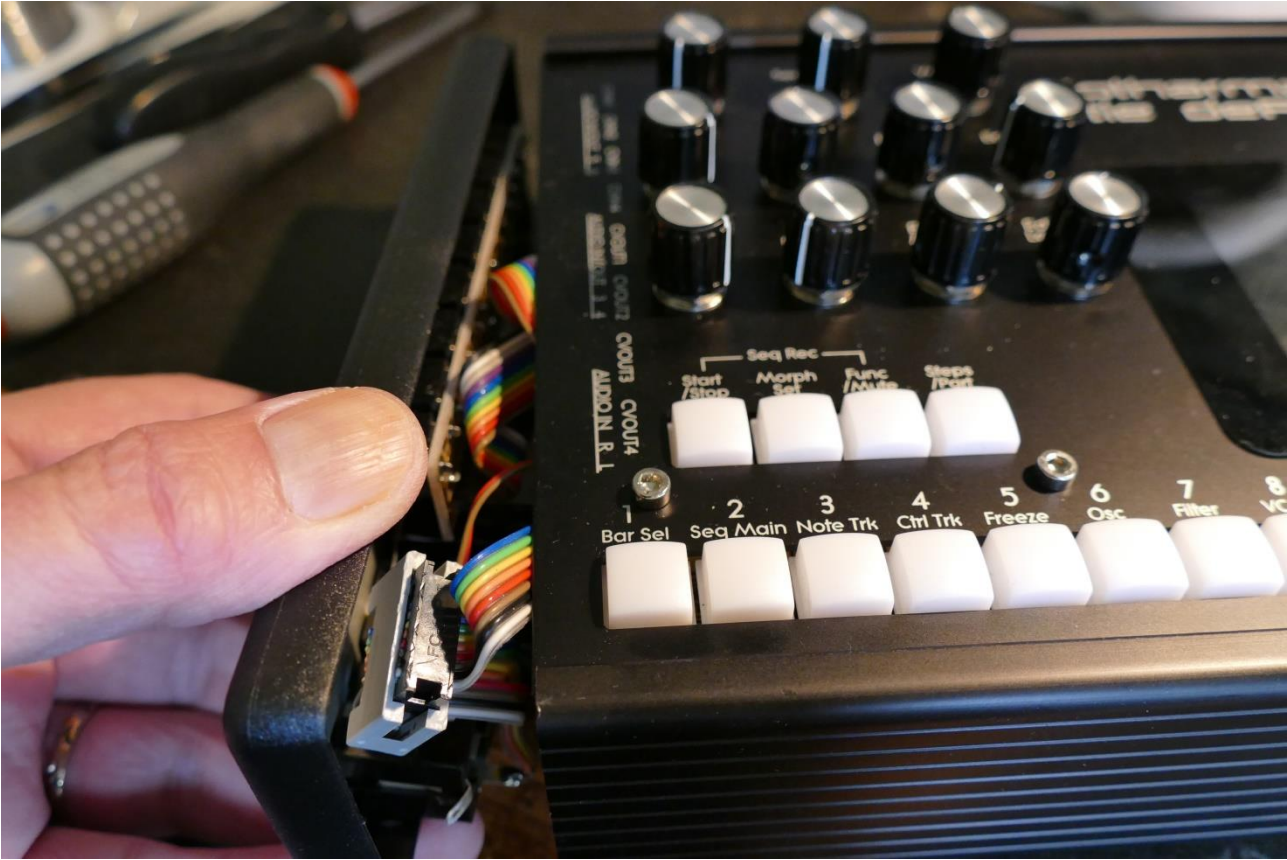
Before you open your LD3, insert the power supply in a wall socket, and touch the metal on the connector. This will unload you from any static electricity, that might damage the electronics.



Remove the 4 screws in the corners of the left end panel, using the Phillips screwdriver.



Take the left end panel a bit out.



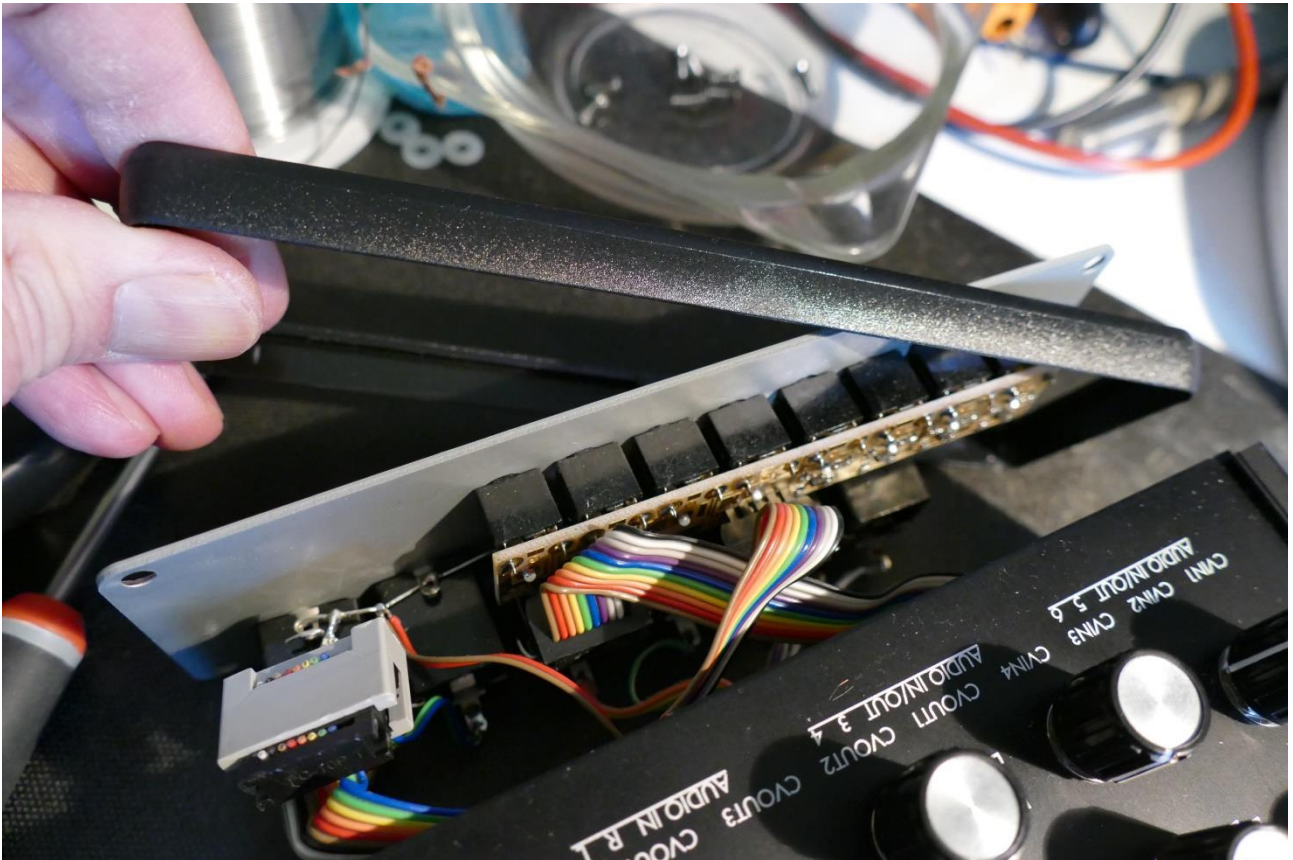
Turn your LD3 around, and remove the 4 screws in the corners of the right end panel, again using the Phillips screwdriver.



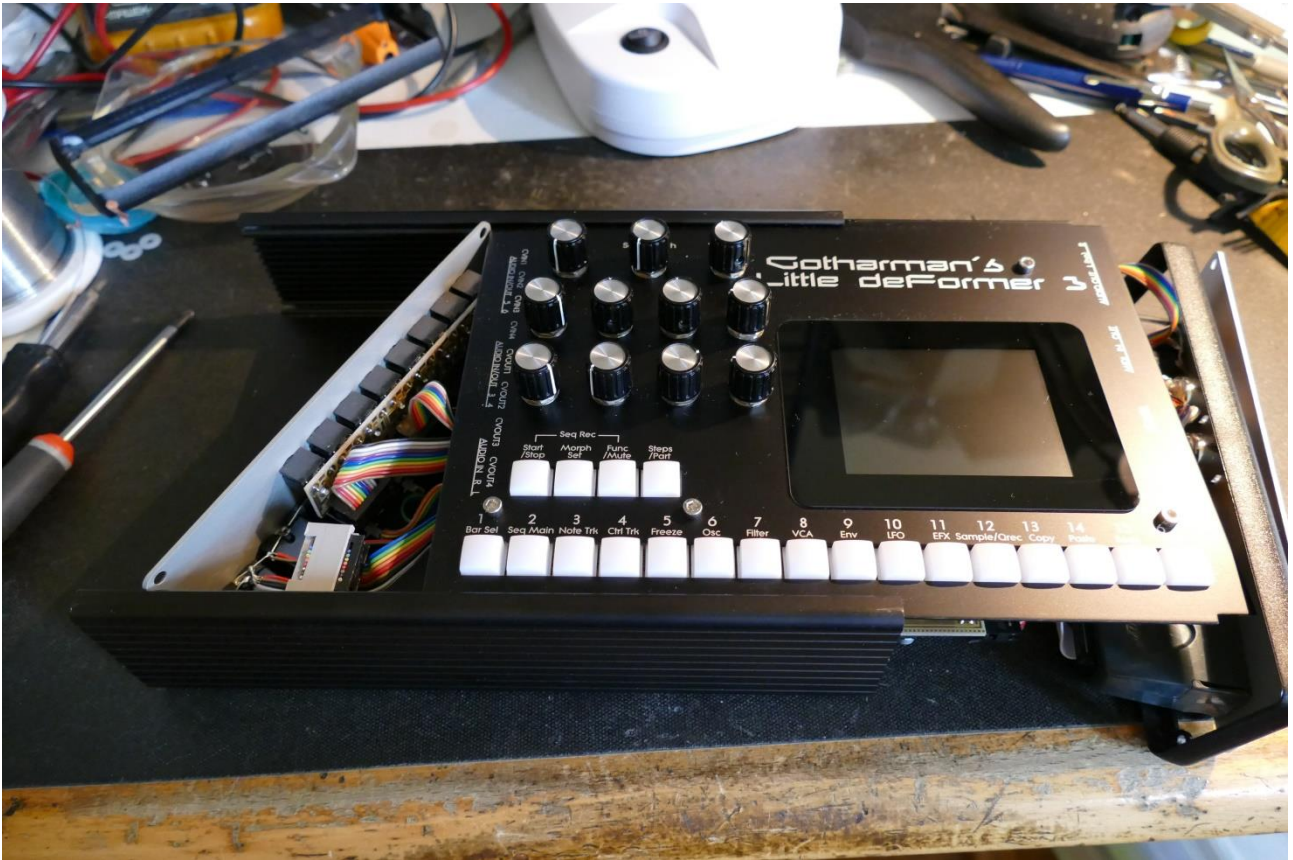
Take the right end panel a bit out.



Remove the plastic frame from the left end panel.



Now carefully slide the front panel towards the right, while navigating the left end panel, to stay inside the box, in order not to scratch the top of the box.

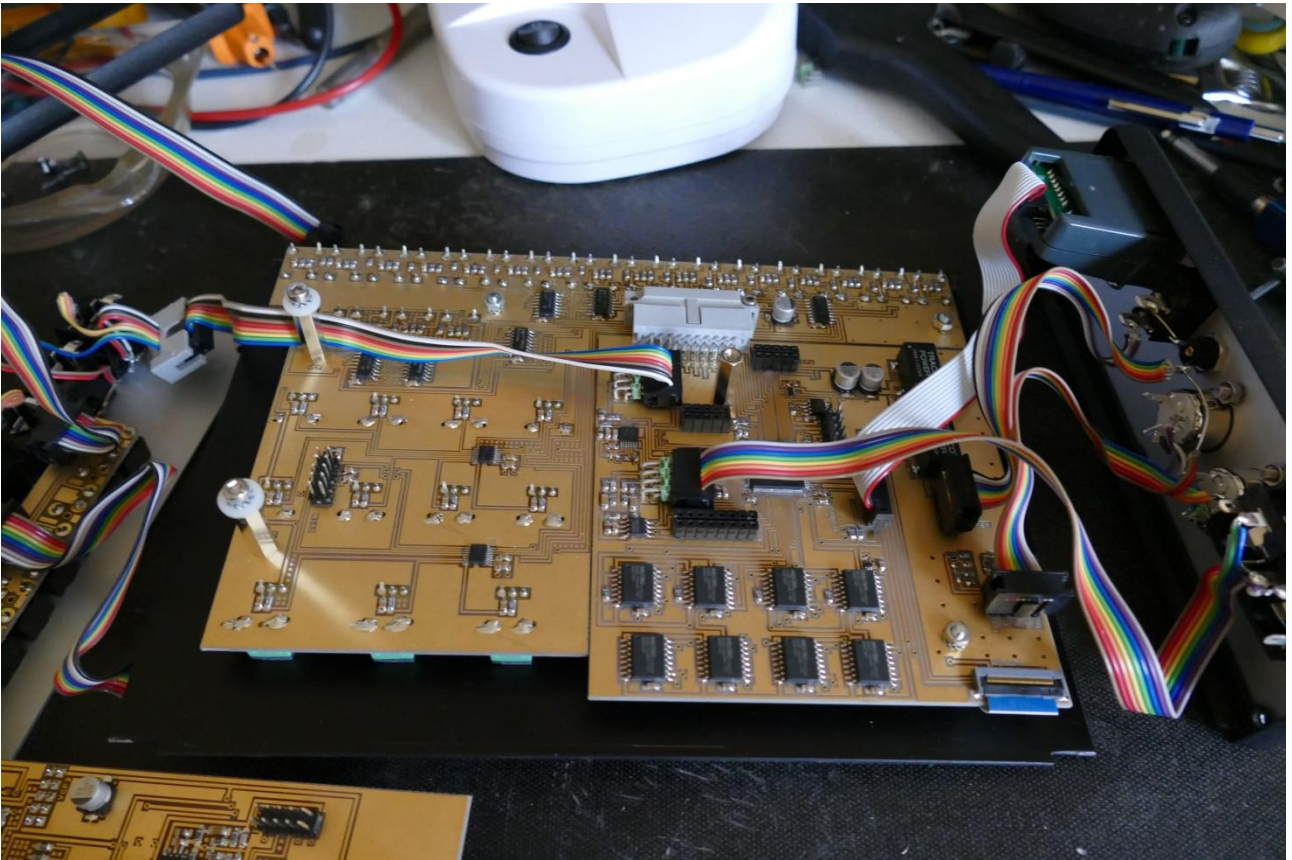


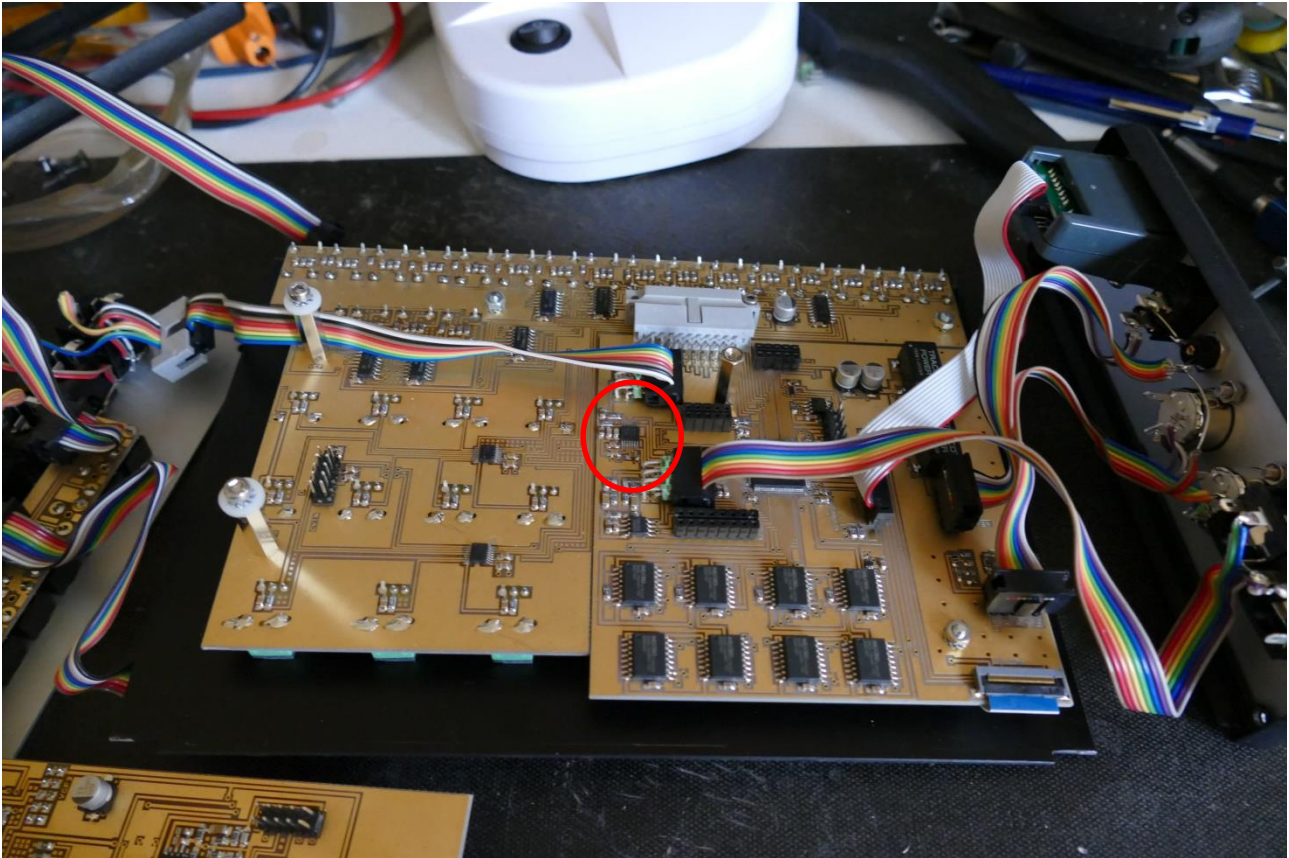
Slide the front panel all the way to the right, so that it is free of the box.

Put the box somewhere else, so you have the front panel and the two end panels for themselves.

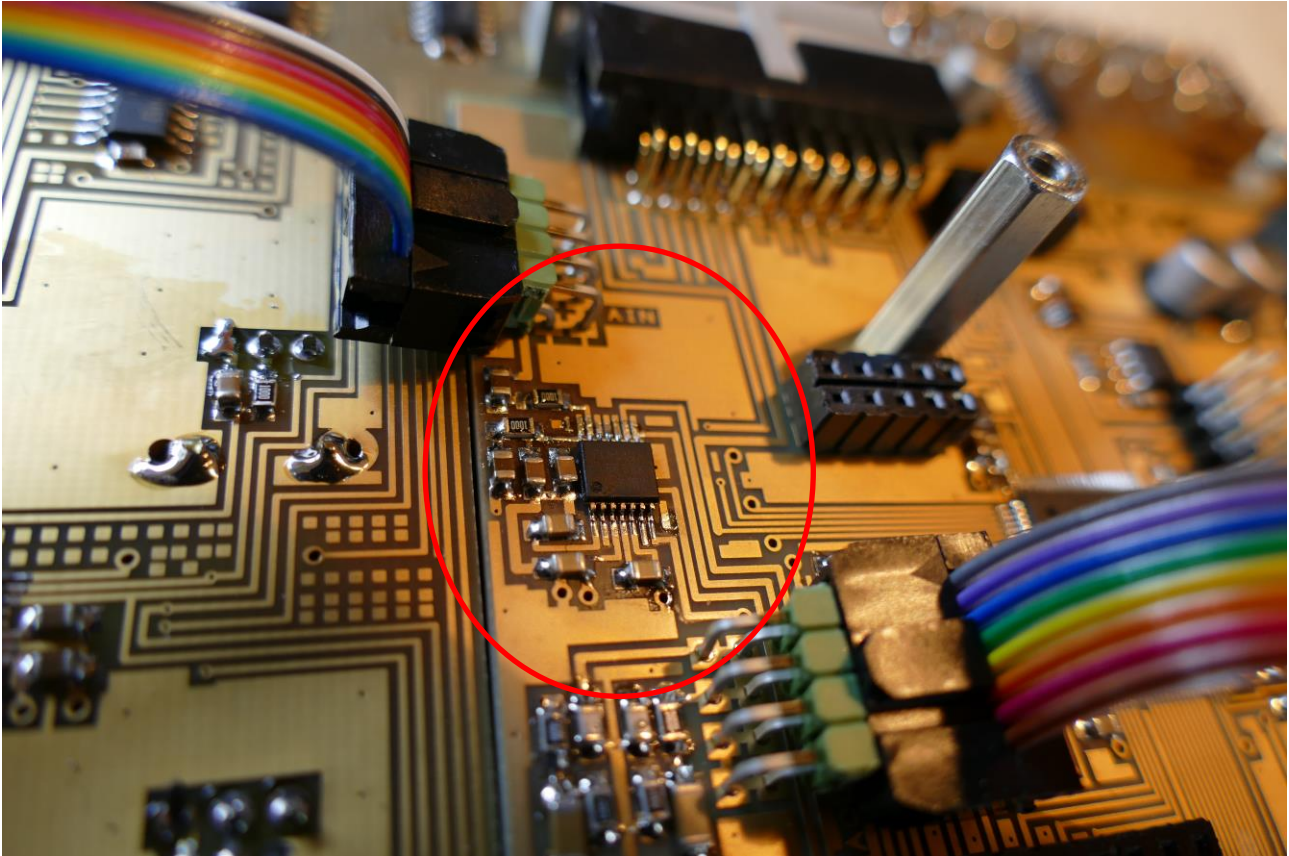


Flip LD3 around, so that the front panel is facing downwards.



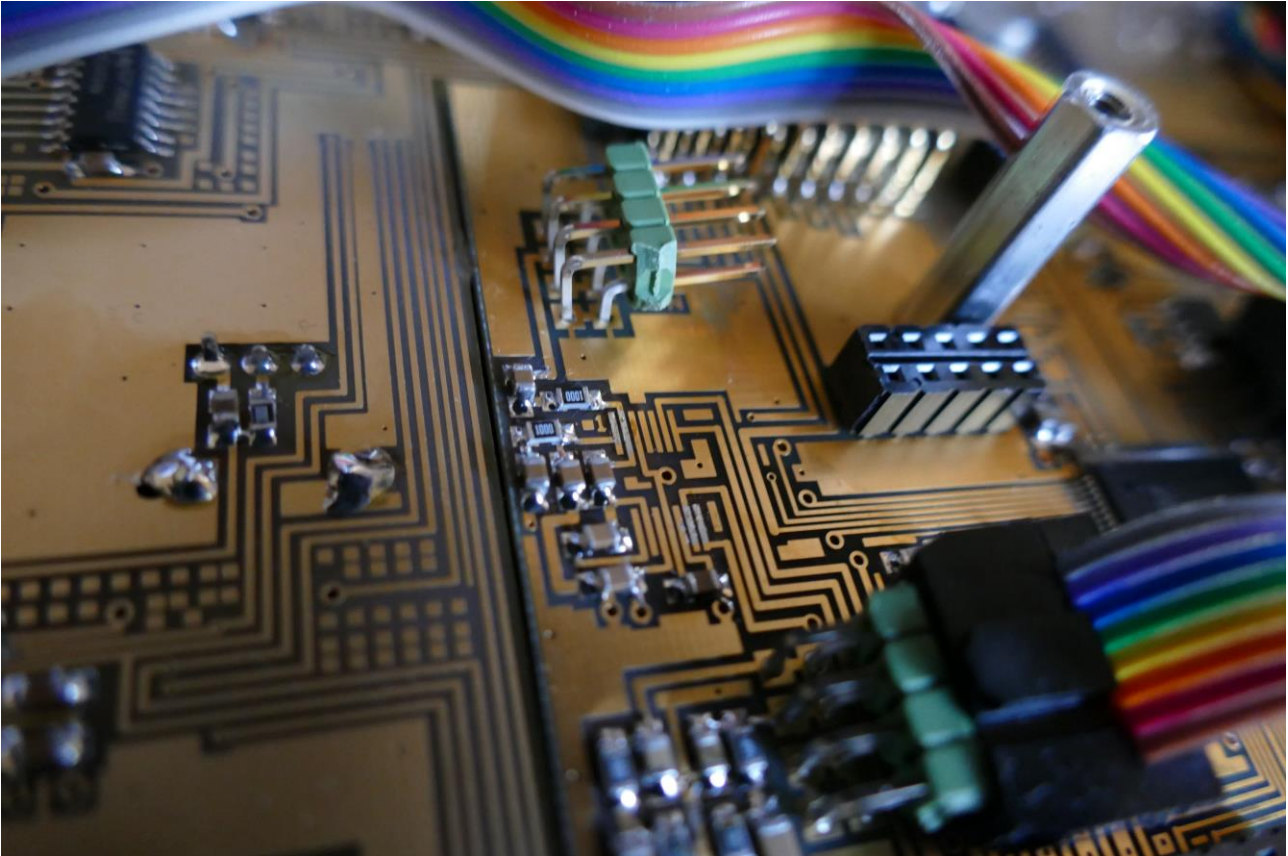


Now you will need to remove an SMT audio converter, that is placed on the LD3 mainboard. Its location is marked with a red circle on the picture.



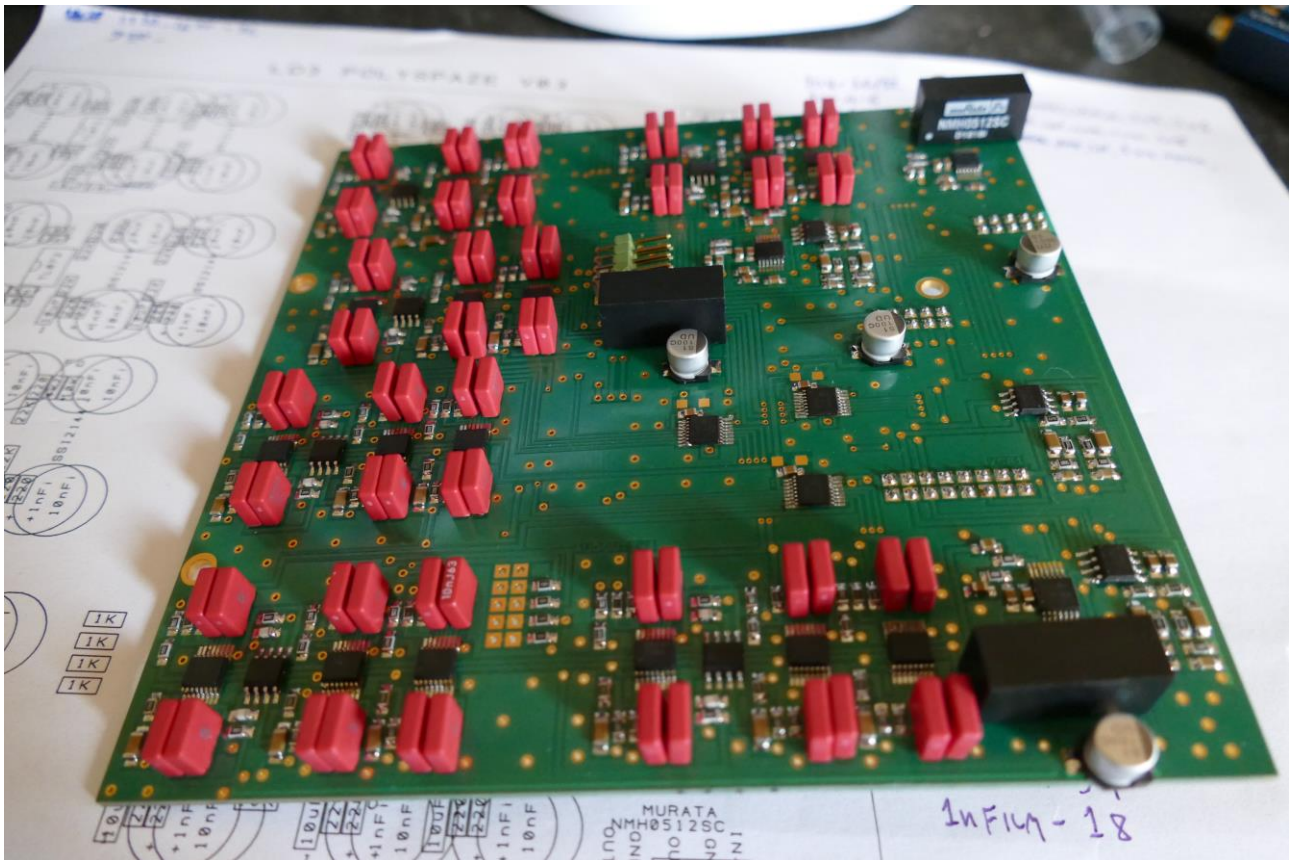
A closer look at the audio converter, that MUST be removed, before installing Spazeboard6.

I use Chip Quik to remove such things, but you should use the tools, that you are comfortable with.

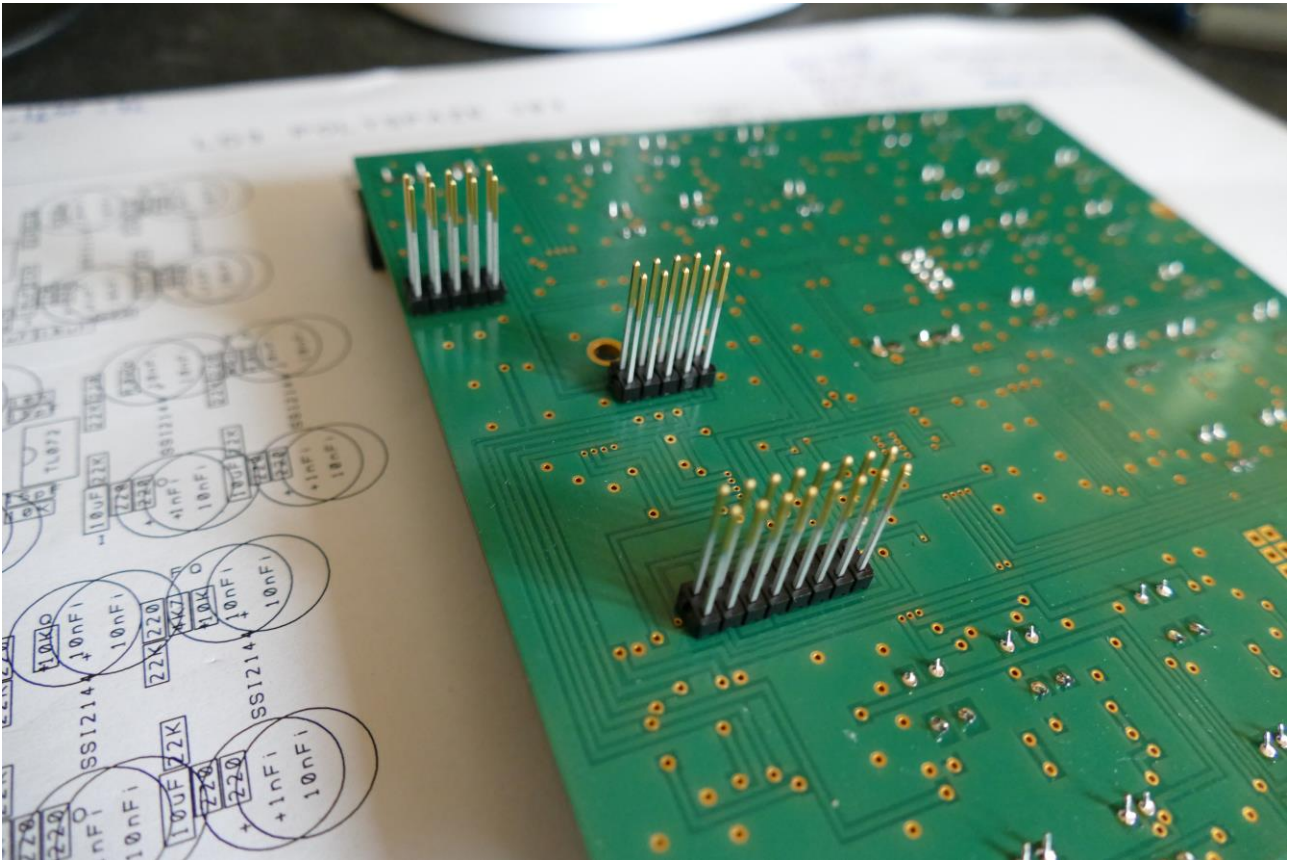


After the audio converter has been removed, make sure to clean the area from solder rests, using solder wick and isopropyl alcohol. Make sure, that there are no short-circuits.

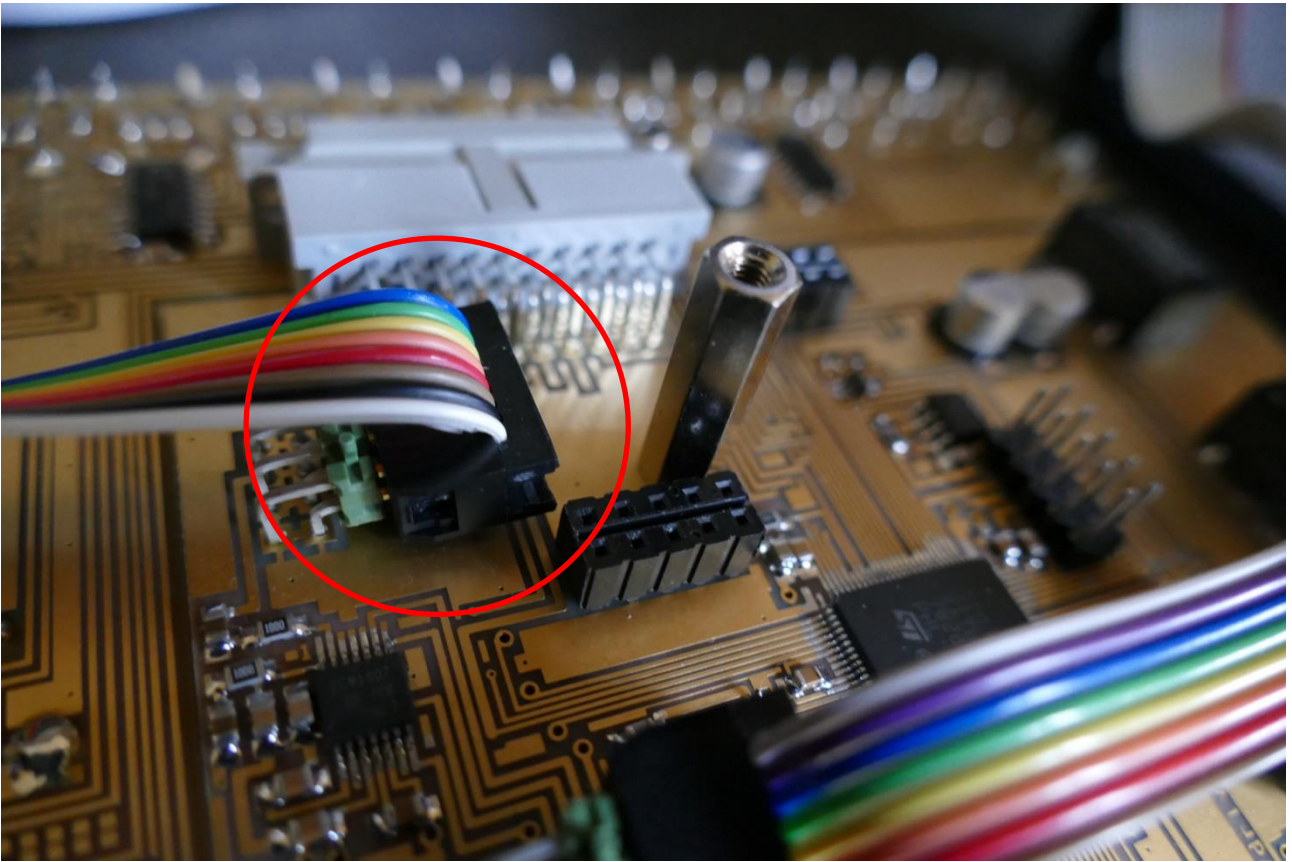
Now it's time to bring in the Spazeboard itself. Spazeboard6 looks like this:



If you turn it upside down, it will look like this:



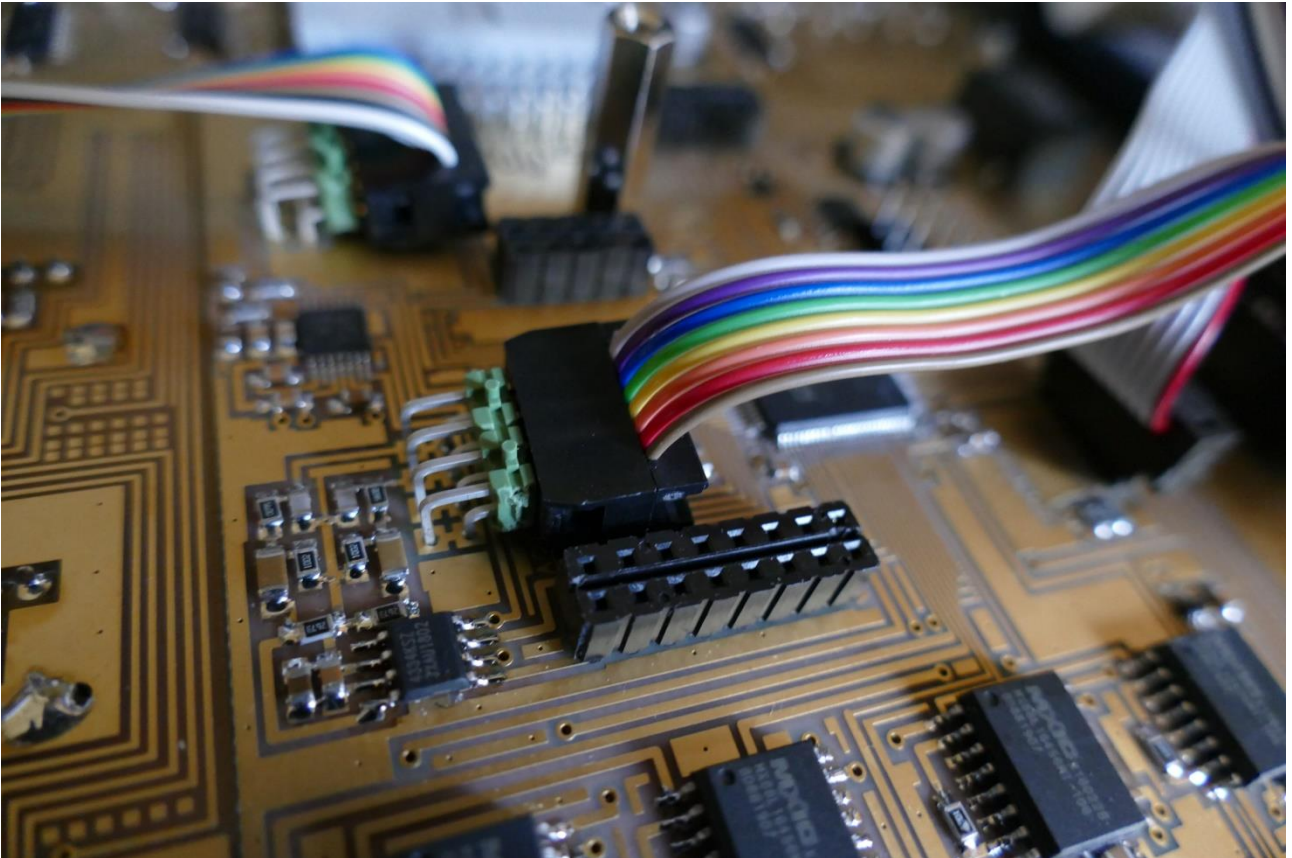
The 3 connectors on the bottom, you will have to match with the 3 connectors on the LD3 mainboard.



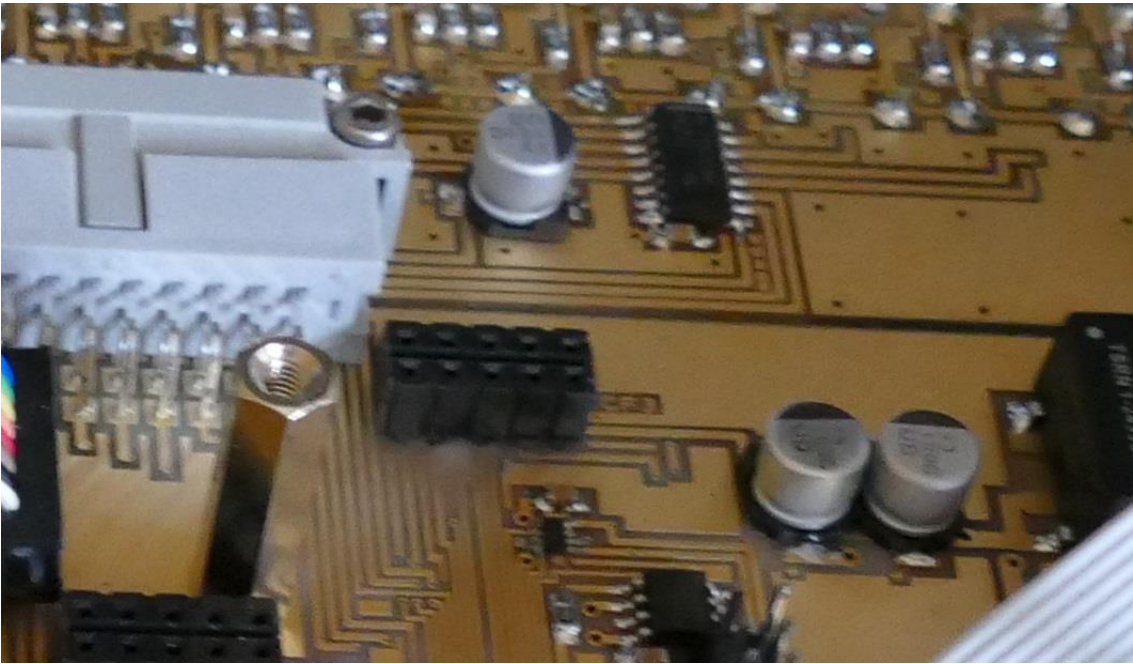
If a screw is located on the spacer on the main board, remove this, so that the spacer looks like on the picture.

Please notice the connector near the spacer. This is one of the connectors, that is needed for Spazeboard6.

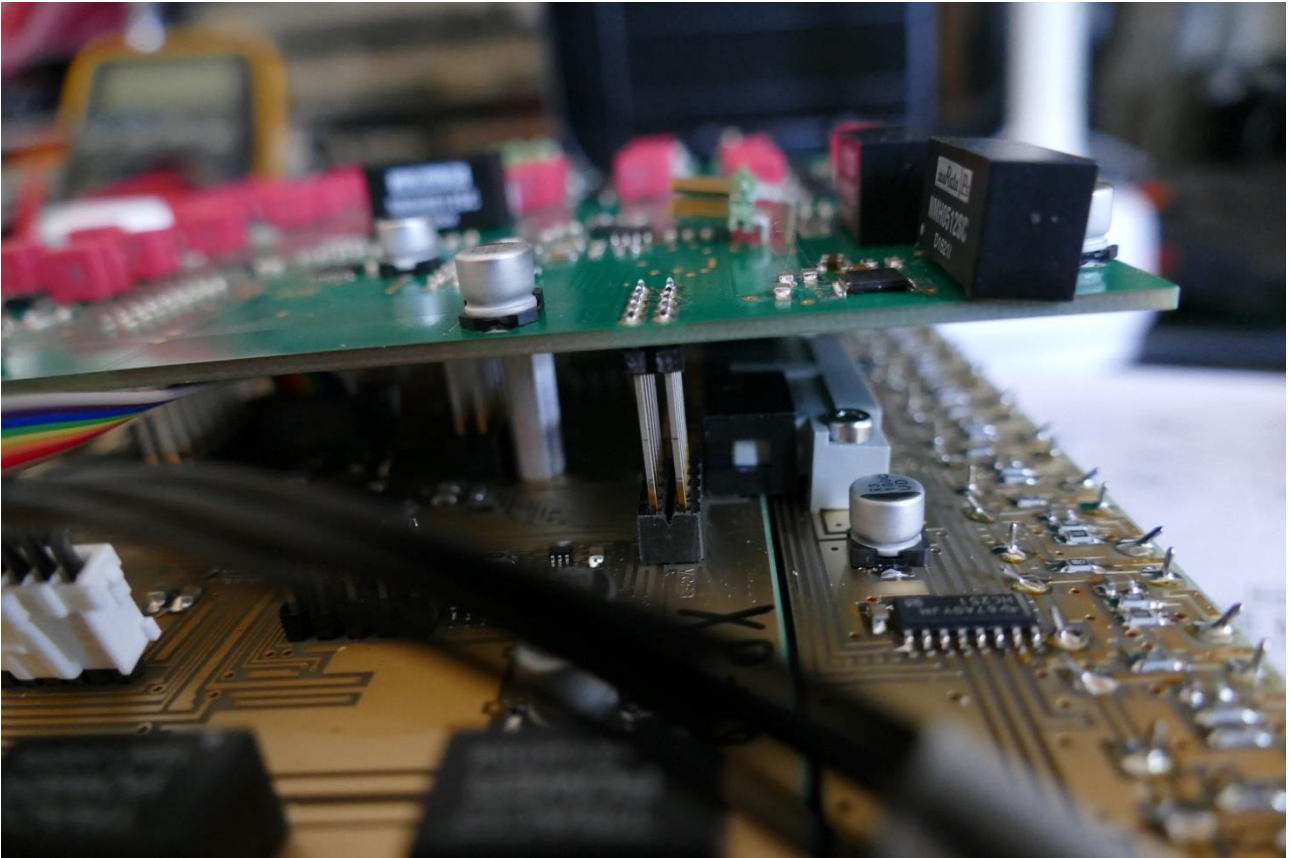
Before you install Spazeboard6, the audio inputs cable, marked with a red circle, must be removed. Since the audio converter has been removed, this has no function anyway. It must be connected to Spazeboard6 instead, for the audio inputs to go through VCF5 and 6.



Near the cable, connected to the main audio outputs, are another connector, needed for Spazeboard6.

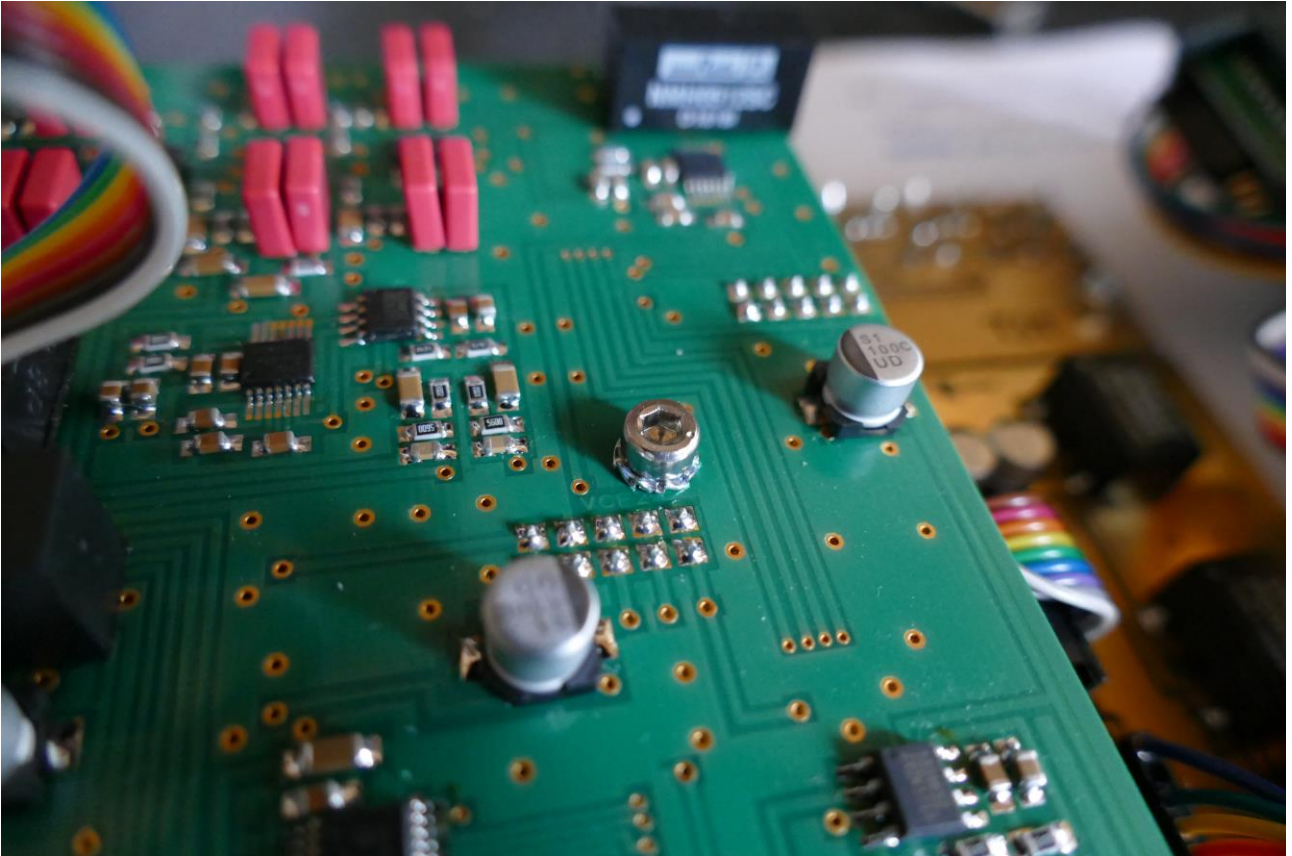


Near the connector to the knob board, are the third connector, needed for Spazeboard6.

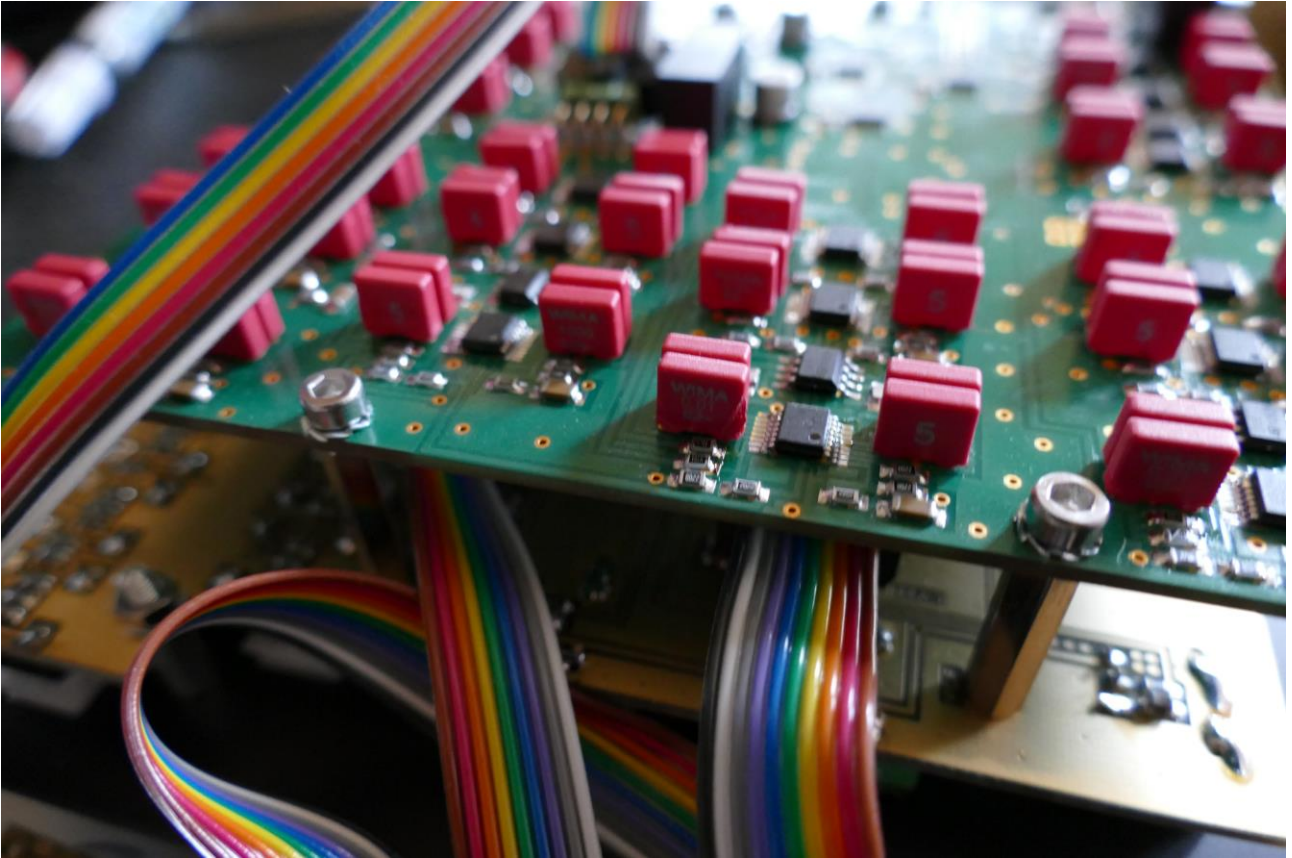


Now carefully attach Spazeboard6 to the LD3 main board. Make sure that the connectors on Spazeboard6 goes perfectly into the connectors on the main board.

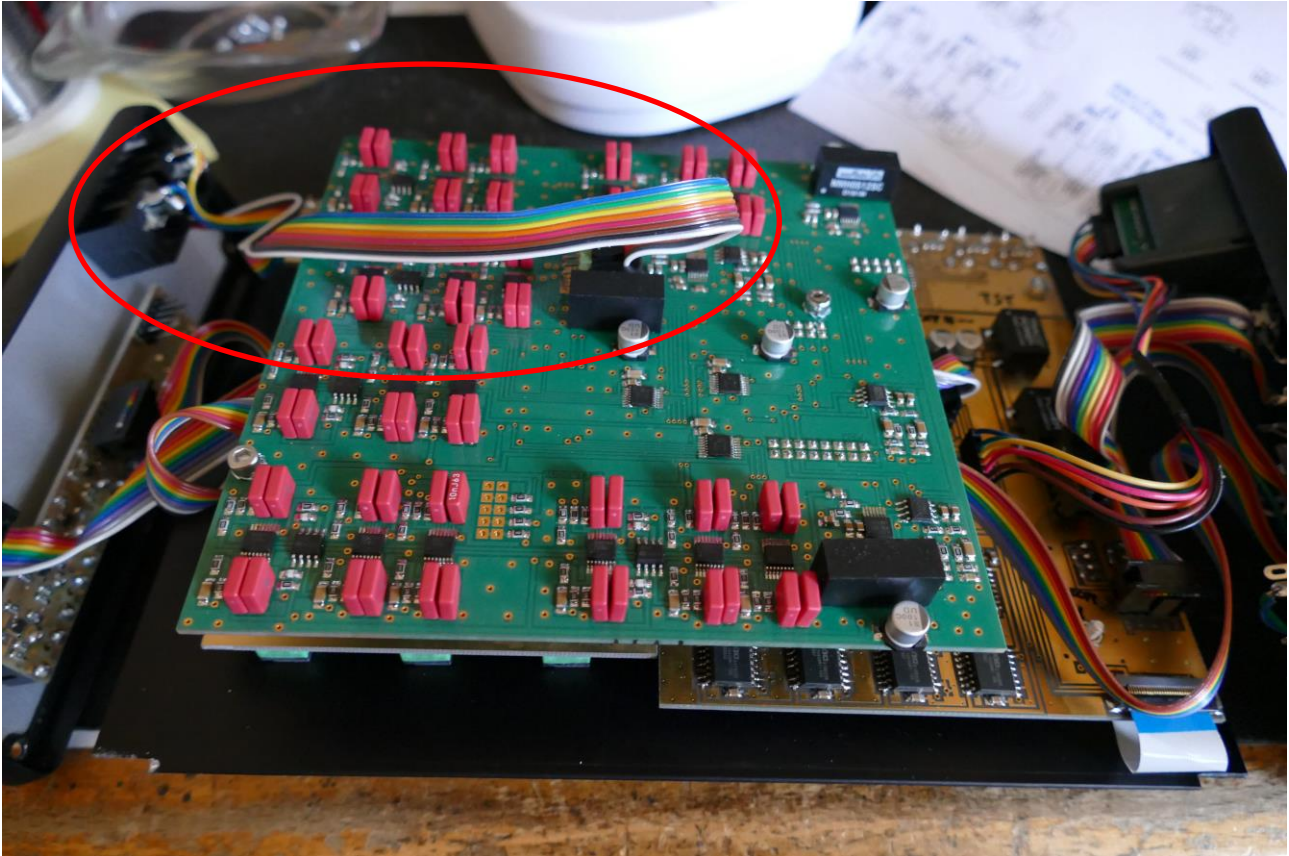
After making sure, that the connectors fits each other perfectly, press Spazeboard6 into these. Check an extra time, that the connectors fits. **THIS IS VERY IMPORTANT!**



Now, you will have to mount 3 hex screws with washers.
The first one is located on the right side of Spazeboard6.



The 2 other screws with washers must be mounted on the left side of Spazeboard6.



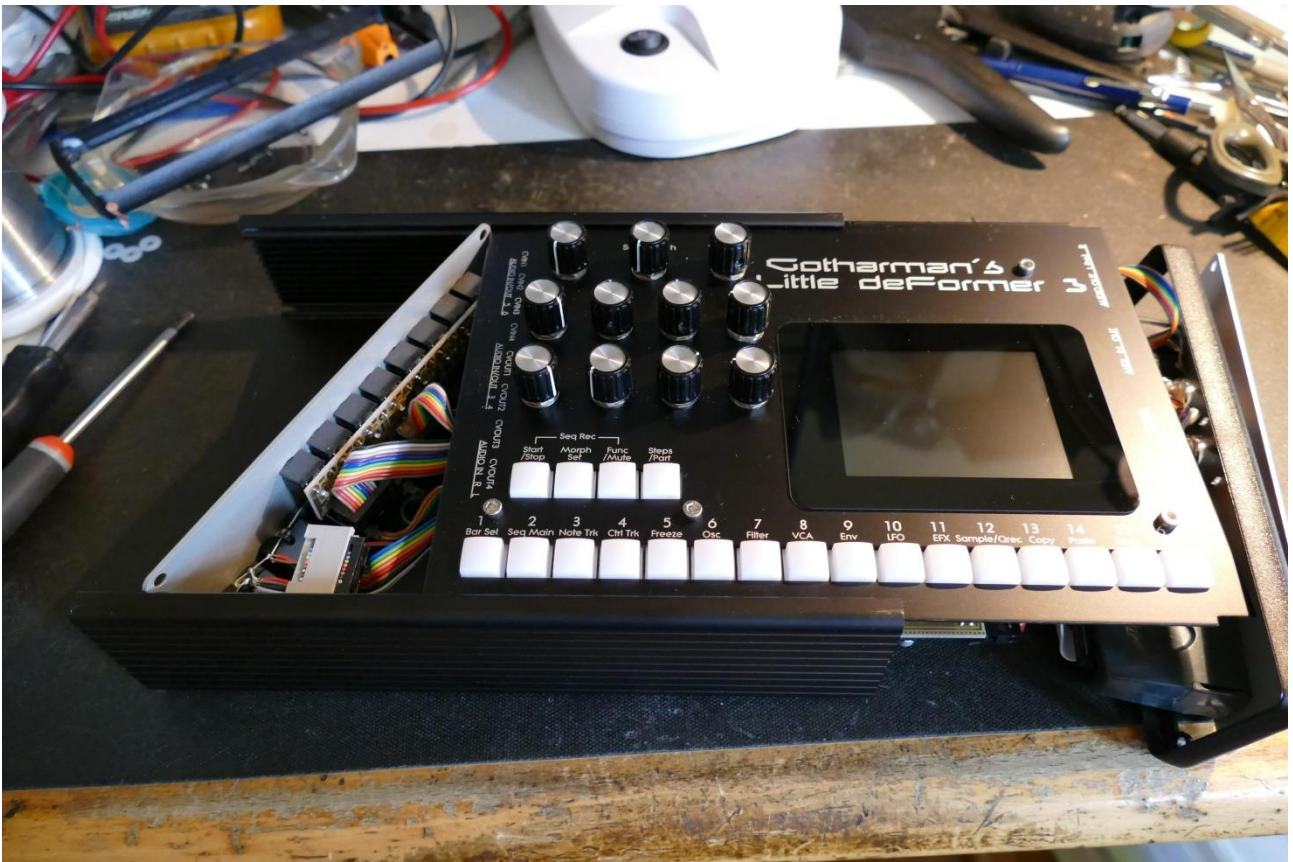
Now mount the cable from the audio inputs, to the connector on Spazeboard6.

You have now successfully installed Spazeboard6!
Now it is time to put your LD3 back together again.

Turn it over, so the front panel is facing upwards again.



Slide the front panel into the box, from the right, towards left. Again, hold the left end panel, so that it stays inside the box, to prevent scratching the top of the box.



Put the plastic frame back on the left end panel.

Put the right end panel in place. Take care that none of the cables attached to it, are squeezed between the panel and the box.

Put the 4 screws back, and tighten them.

Put the left end panel in place. . Take care that none of the cables attached to it, are squeezed between the panel and the box, or the panel and Spazeboard6.

Put the 4 screws back, and tighten them.

PLEASE READ THE NEXT PAGE, BEFORE YOU ENJOY SPAZEBOARD6!

Installing the Spazeboard6 firmware

Even if you have already installed the Spazeboard6 firmware, you will need to do this again.

In the Spazeboard6 firmware available on the Gotharman website, all the analog functions are locked. This is done, in order to protect the hardware, so it is possible to try out the Spazeboard6 firmware, without having to remove the audio inputs converter. If the analog functions had not been locked, it would cause the audio input converter to malfunction.

So now you should turn on your LD3, and install the Spazeboard6 firmware, that you received by email, after your purchase, using the USB Update function.

PLEASE NOTICE: The analog functions will only be un-locked, on the LD3 that has the serial number, that you added when placing your Spazeboard6 order.

Now you can finally enjoy your Spazeboard6!

Written by:
Flemming Christensen
Gotharman
2019