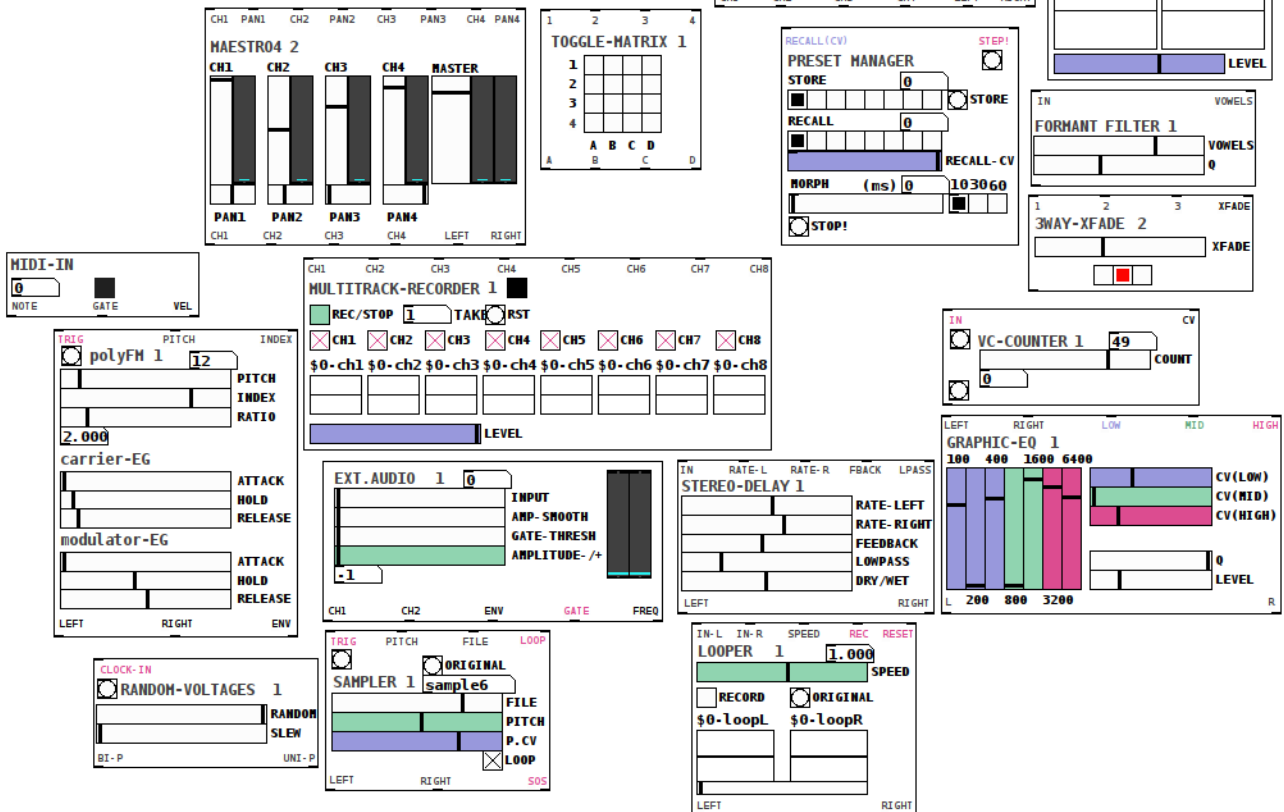


AUTOMATONISM



Automatonism v2.1

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Documentation réalisée à partir des patches d'aide.
par Les Portes Logiques, Quimper, déc. 2018
<http://lesporteslogiques.net>

welcome

Welcome_to_AUTOMATONISM:_Making_Music_with_Self-playing_Machines

FOLLOW THESE SIMPLE STEPS TO GET STARTED:

1 When starting the AUTOMATONISM you'll see a list of all current available modules to the right. Click the button to the left of the module you want to load and it will appear to the left in the parent window. The module list will then disappear. To call back the module list you can click the grey link in the upper left called "MODULES(esc)" or simply upress "esc" on your keyboard

2 Create a BASIC-OSC from the module list.

3 Open the module list and create the module in the lower right corner called MAESTRO4(dac-). This is the digital-to-analog-converter and is one of the modules that passes on sound to your speakers. Now you can soon make a sound.

4 Pure Data has an EDIT MODE and a PERFORMANCE MODE. To make patch connections between objects we need to be in EDIT MODE and to interact with the GUI we need to be in PERFORMANCE MODE. Toggle between the two modes with cmd+e.

5 All inputs are at the top of the modules and all outputs are at the bottom. In EDIT MODE, connect the BASIC-OSC's output CH1 on the MAESTRO4. Now, you should have sound. Switch to PERFORMANCE MODE and use the mouse to move the PITCH slider on the BASIC-OSC module.

6 Right click on any module and choose HELP to read more about a module's functions.

7 To save parameter states on each module, you have to save the pd file, like you normally would do from the FILE menu or cmd+s, and then click the grey SAVE! canvas in the left bottom corner(or simply press ENTER on your keyboard) and you'll see the canvas change color from grey to green for a brief moment. All parameter values are now written to textfiles and can be loaded to the equivalent module and instance number next time the patch is loaded.

8 You can call modules without using the module list simply by creating a new object in pd and type the module's name. For the state saving mechanism to work properly you do need to give the module a unique numerical argument. Type "basic-osc 1" for an oscillator and "basic-osc 2" if you need another one. By using the MODULES list this unique numbering is done automatically for you. Use the RESET link in the upper left corner when starting a new patch as this will reset the counter of unique numbers for modules.

9 IMPORTANT!!! In order for Automatonism to work properly, each patch needs to contain the correct folder structure. A folder--that you can name freely--must contain the folder "patch_editor_abs" and the pd-file "main.pd". Copy the entire main parent folder to start a new project. Please do not add the abstractions to your PD Search path, because the state saving system will be writing textfiles in your external folders.

ABOUT RESET BUTTON: Only use the RESET button when your canvas is empty. If you click it in the middle of a patch, modules will start getting identical state saving numbers to already existing modules. In other words, if you click it in the middle of a patch, state saving WILL NOT WORK! The RESET button is there for when you start over, and perhaps do not want to create "BASIC-OSC 99", but start from zero for a clearer view of the patch.

welcome_menu

WELCOME-TO-THE-AUTOMATONISM-MANUAL!

SIGNALFLOW

COLOR CODE

STATE SAVING

HELP FILES

FOLDER HIERARCHY

FEEDBACK-&-DSP-LOOPS

LICENSE

SIGNAL-FLOW

There are three types of signals in the patch editor: audio, cv and trigger/gate/pulse/clock. The latter are marked with PINK inlets and outlets. General rule in the patch editor is that all connections are possible. Or, at least, just like in any hardware modular system - all connections might not work but there's no harm in trying. !! The patch-editor uses color-coding to clarify the signal flow as much as possible.



Purple sliders and inlets means the slider will function as an attenuator for incoming CV.



Green sliders and inlets tells you that the parameter is bi-polar, most oftenly working as an attenuverter for incoming CV.

CLOCK-IN

And pink inlets means the input is expecting a trigger/gate signal to function properly.

All inputs and outputs in the editor are at signal rate within the range of -1 to 1 In contrast to normal pd-programming there is no distinction between control rate and signal rate messages. Everything is converted to signal rate.

How to convert a simple bang to a trigger/gate signal in the patch editor:

```
bang
|
| 0, 0 1 1|
|
vline~
```

How to convert back to a control rate bang:

```
bang
|
| 1 0, 0 1 1|
|
vline~
|
threshold~ 0.99 0 1 0
|
bang
```

How convert control numbers to the patch editor's signal flow:

```
0-127
|
| 127
|
sig~
```



Purple sliders and inlets means the slider will function as an attenuator for incoming CV.



Green sliders and inlets tells you that the parameter is bi-polar, most oftenly working as an attenuverter for incoming CV.

CLOCK-IN

And pink inlets means the input is expecting a trigger/gate signal or short impulse(square-wave) to function properly.

STATE_SAVING

The state saving system works by writing parameter values to textfiles in the folder "statesave" inside the "patch editor abs" folder. When you load a module from the module list it is automatically being given a unique creation argument which creates a textfile unique for that instance of the module. To save your patch, simply save as you normally would from the FILE menu and the press ENTER on your keyboard or click the SAVE button in the upper left corner of the main window.

IMPORTANT!!! To start a new project, the entire folder structure needs to be copied since the abstractions and textfiles are unique to every project. Click the RESET button in the upper left corner to reset the counter that gives unique creation arguments for the modules.

HELP-FILES

Right + click on individual modules and choose "HELP" to access detailed help-files on each modules inlets, outlets and functionalities.

FEEDBACK-&-DSP-LOOPS

Feedback patching is very much possible in the AUTOMATONISM, just a little bit differently from physical modular systems. If you try to patch a module into itself or any kind of feedback routing, Pure Data will become silent and you'll get a "DSP loop detected" message in the Pd window. To avoid this you need to delay the signal a tiny tiny bit. Normal [send~] and [receive~] objects automatically does this in Pd. Send the signal to be feedbacked into a [send~ MyFeedback] and pick it up using [receive~ MyFeedback] and then patch into the input to close the feedback loop. Obviously "MyFeedback" could be anything and the abbreviation for send and receive [s~] and [r~] work equally as well.

FOLDER-HIEARCHY

Inside your project folder you'll find a file called "main.pd" and a folder called "patch editor abs". The "main.pd" must stay named the same for the playground to function properly. Also, each of your projects/patches needs to be in a separate folder with its own "main.pd" and "patch_editor_abs" folder. Copy & paste the project folder, which you can give any name, when you want to start a new project/patch.

LICENSE

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modules

WELCOME-TO-AUTOMATONISM		
MODULES		
(New modules for Automatonism v2.0 have GREEN buttons)		
OSCILLATORS:	SOUND-PROCESSORS:	UTILITY&LOGIC:
<input type="checkbox"/> <- BASIC-OSC	<input type="checkbox"/> <- LP-FILTER	<input type="checkbox"/> <- CLOCK
<input type="checkbox"/> <- BWL-OSC	<input type="checkbox"/> <- HP-FILTER	<input type="checkbox"/> <- CLOCK-MULTIPLY
<input type="checkbox"/> <- WTABLE	<input type="checkbox"/> <- BP-FILTER	<input type="checkbox"/> <- CLOCK-DIVIDER-EVEN
<input type="checkbox"/> <- KARPLUS	<input type="checkbox"/> <- LADDER-FILTER	<input type="checkbox"/> <- CLOCK-DIVIDER-ODD
<input type="checkbox"/> <- 20PFH	<input type="checkbox"/> <- ANALOG-FILTER	<input type="checkbox"/> <- VC-COUNTER
<input type="checkbox"/> <- TRIGONS	<input type="checkbox"/> <- FORMANT	<input type="checkbox"/> <- TRIGGER-DELAY
<input type="checkbox"/> <- TRIAD	<input type="checkbox"/> <- BITCRUSH	<input checked="" type="checkbox"/> <- TRIGGER-TRAIN
<input checked="" type="checkbox"/> <- PD-303	<input type="checkbox"/> <- WAVEFOLD	<input type="checkbox"/> <- INVERTER
<input checked="" type="checkbox"/> <- POLYSYNTH	<input type="checkbox"/> <- PHASOR	<input type="checkbox"/> <- SCOPE
<input checked="" type="checkbox"/> <- POLYFM	<input type="checkbox"/> <- COMBFILTER	<input type="checkbox"/> <- SLEW
<input type="checkbox"/> <- KICK	<input type="checkbox"/> <- GRANULAR-DELAY	<input type="checkbox"/> <- PROBABILITY
<input type="checkbox"/> <- SNARE	<input type="checkbox"/> <- MILLERVERB	<input type="checkbox"/> <- QUANTIZER
<input type="checkbox"/> <- NOISE	<input type="checkbox"/> <- STEREO-DELAY	<input type="checkbox"/> <- 401-SWITCH
<input checked="" type="checkbox"/> <- LOOPER	<input type="checkbox"/> <- CHORUS	<input type="checkbox"/> <- IT04-SWITCH
<input checked="" type="checkbox"/> <- SAMPLER	<input checked="" type="checkbox"/> <- OVERDRIVE	<input type="checkbox"/> <- ENV-FOLLOWER
MODULATION:	<input checked="" type="checkbox"/> <- GRAPHIC-EQ	<input type="checkbox"/> <- PRESET-MANAGER
<input type="checkbox"/> <- BASIC-LFO	<input checked="" type="checkbox"/> <- MEGAVIB	<input type="checkbox"/> <- PARAM-MUDGE
<input type="checkbox"/> <- SINEBANK	<input checked="" type="checkbox"/> <- PITCHSHIFTER	<input checked="" type="checkbox"/> <- AUDIO-RECORDER
<input type="checkbox"/> <- ATTACK-HOLD-RELEASE(ahr)	VCA/MIXERS:	<input checked="" type="checkbox"/> <- MULTITRACK-RECORDER
<input type="checkbox"/> <- DECAY	<input type="checkbox"/> <- VCA	<input checked="" type="checkbox"/> <- EXTERNAL-AUDIO
<input type="checkbox"/> <- SLOPE(function-generator)	<input type="checkbox"/> <- LPG(lopass-gate)	<input checked="" type="checkbox"/> <- MANUAL-TRIGGERS
<input type="checkbox"/> <- ADSR	<input type="checkbox"/> <- XFADE	MIDI:
<input type="checkbox"/> <- RANDOM-GATES	<input type="checkbox"/> <- 3WAY-XFADE	<input type="checkbox"/> <- MIDI-IN
<input type="checkbox"/> <- RANDOM-VOLTAGES	<input type="checkbox"/> <- POLARIZER	<input type="checkbox"/> <- MIDI-OUT
<input type="checkbox"/> <- S&H(sample&hold)	<input type="checkbox"/> <- MIXER2(2-channel)	ORGANELLE:
SEQUENCERS:	<input type="checkbox"/> <- MIXER4(4-channel)	<input type="checkbox"/> <- ORGANELLE-KNOBS
<input type="checkbox"/> <- 8STEPS	<input type="checkbox"/> <- TOGGLE-MATRIX	<input type="checkbox"/> <- ORGANELLE-KEYBOARD!
<input type="checkbox"/> <- MEGA-SEQUENCER!	<input type="checkbox"/> <- MAESTRO4(dac-)	<input type="checkbox"/> <- ORGANELLE-AUX
<input type="checkbox"/> <- TRIGGER-SEQ		<input type="checkbox"/> <- ORGANELLE-OUTPUT
<input type="checkbox"/> <- GATE-SEQ		

basic_oscillator

BASIC-OSCILLATOR

The "BASIC-OSC" module is a simple oscillator providing sine, triangle, saw and pulsewave with adjustable pulsewidth. The Basic Oscillator is crude and digital--it sounds best in lower registers but is low on the CPU. For less aliasing and digital artefacts use "BWL-OSC".

CONTROLS:

PITCH - controls the pitch of the oscillator within the range of 8.175Hz -12543Hz (midi notes 0-127)

P-WIDTH - controls the width of the pulsewave

PWM - attenuator for incoming cv at the PWM input

FM - attenuator for incoming CV at the FM input.

WAVESELECTOR - switch between sine, triangle, saw and pulsewave

INPUTS:

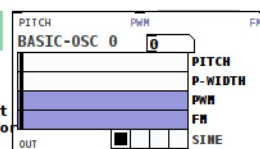
PITCH - input for controlling the pitch with CV

PWM - input for controlling the pulsewidth of the pulsewave with CV

FM - frequency modulation input with onboard attenuator

OUTPUTS:

OUT - output



bandwidth_limited_oscillator

BANDWIDTH-LIMITED-OSCILLATOR

The "BWL-OSC" module is a bandwidth-limited oscillator providing sine, triangle, saw and pulsewave. A waveshaper circuit adds harmonics to the sine and triangle, produces a detuned supersaw and pulsewidth modulation.

CONTROLS:

PITCH - controls the pitch of the oscillator within the range of 8.175Hz -12543Hz (midi notes 0-127)

WSHAPE - controls sine & triangle harmonics, saw detuning and pulsewidth.

CV - attenuator for incoming CV at WSHAPE input.

FM - attenuator for incoming CV at the FM input.

WAVESELECTOR - switch between sine, triangle, saw and pulsewave

INPUTS:

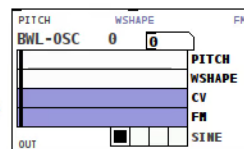
PITCH - input for controlling the pitch with CV

WSHAPE - input for controlling the WSHAPE parameter with CV

FM - frequency modulation input with onboard attenuator

OUTPUTS:

OUT - output



wavetable_oscillator

WAVETABLE-OSCILLATOR:

The "wtable" module is a three dimensional wavetable oscillator. Scan, morph and crossfade 18 wavetables with Y & Z.

CONTROLS:

PITCH - controls the pitch of the oscillator within the range of 8.175Hz -12543Hz (midi notes 0-127)

X - morph between three tables

Y - morph between banks of tables

Z - morph between

FM - attenuator for incoming CV at the FM input.

INPUTS:

PITCH - input for controlling the pitch with CV

X - input for cv-control over the X-axis

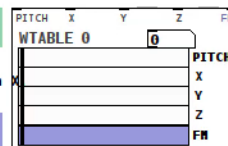
Y - input for cv-control over the Y-axis

Z - input for cv-control over the Z-axis

FM - frequency modulation input with onboard attenuator

OUTPUTS:

OUT - output



karplus_strong_resonator

KARPLUS-STRONG-RESONATOR:

The "karplus" module is an adaptation of the karplus strong algorithm that models the sound of resonating strings. The module does not produce constant sound like other oscillators but needs a trigger/gate/pulse signal at the EXC. input to trigger an internal envelope and make a sound. White noise is used to trigger a short impulse/burst to trigger the resonator. You can remove the white noise by patching any sound/signal into the EXT.EX input and use that as the exciter sound source instead. KARPLUS is actually four independent resonators, which are distributed to left and right outputs to produce a stereo polyphonic feel.

CONTROLS:

PITCH - controls the pitch of the resonating strings.

CONTOUR - control the envelope shape of the noise exciter

DAMP - damp the strings

RESONATE - controls the decay length of the resonator.

COLOR - a simple lowpass filter on the output

INPUTS:

EXC. - trigger/gate input to trigger the resonator

EXT.EX - remove the normalization of white noise and use this external source for excitation instead.

PITCH - input for controlling the pitch with CV

CNTR - input for controlling the CONTOUR parameter with CV.

DAMP - input for controlling the DAMP parameter with CV.

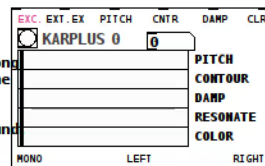
CLR - input for controlling the COLOR parameter with CV.

OUTPUTS:

MONO - output

LEFT - left output

RIGHT - right output



2_operator_fm

2-OPERATOR-FM:

The 2OP-FM module is a 2 operator phase modulation oscillator. Manual + voltage control over ratio and index.

CONTROLS:

CARRIER - controls the pitch of carrier

RATIO - controls the phase modulation ratio.

INDEX - controls the phase modulation index.

CV1(RATIO) - attenuator for incoming CV at the CV1 input

CV2(INDEX) - attenuator for incoming CV at the CV2 input

INPUTS:

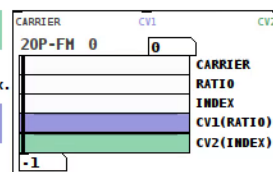
CARRIER - input for controlling the carrier pitch

CV1 - cv input for the **RATIO** control

CV2 - cv input for the **INDEX** control

OUTPUTS:

OUT - output



trigons

TRIGONS:

The "TRIGONS" module is an oscillator with rough edges and sharp triangular spikes. Crossfade between triangular and sawtooth and use an aggressive waveshaper to add harmonics.

CONTROLS:

PITCH - controls the pitch of the oscillator

WAVE - crossfade between triangle wave and sawtooth

HARM. - add harmonics and waveshaping

FM - attenuator for incoming CV at the FM input.

INPUTS:

PITCH - input for controlling the pitch with CV

WAVE - input for cv-control over the WAVE control

HARM. - input for cv-control over the HARM. control

FM - frequency modulation input with onboard attenuator

OUTPUTS:

OUT - output

PITCH	WAVE	HARM.	FM
TRIGONS 0		0	
			PITCH
			WAVE
			HARM.
			FM

triad_oscillator

TRIAD-OSCILLATOR:

The TRIAD oscillator is a duophonic chord module. A trigger/gate at the TRIG input will trigger an internal attack-release envelope and produce a three-note-chord at the output. The core of the module is dual, meaning that the first chord will keep ringing when the next is fired. The chord can be adjusted with the ROOT, INT1 and INT2 sliders. Quantization scale and transpositions are selectable at the bottom of the module.

CONTROLS:

ROOT - select the pitch of the root
INT1 - select the pitch interval 1
INT2 - select the pitch the interval 2
DTUNE - control for slight detuning of the chord
ATTACK - attack time of the internal envelope
RELEASE - release time of the internal envelope
SCALE - select scale
XPOSE - transpose the chord -12 to +12 semitones
WAVESELECTOR- select sawtooth, squarewave or alternate between saw and square

INPUTS:

TRIG - trigger the envelope and the chord
ROOT - cv input for the ROOT
INT1 - cv input for INT1
INT2 - cv input for INT2
XPOSE - cv input for the XPOSE control

OUTPUTS:

TRIGGERED - outputs the chord shaped by the internal envelope
CONSTANT - outputs the chord constantly bypassing the envelope
EG - outputs the envelope to use as a modulation source elsewhere in the system

The screenshot shows the 'TRIAD OSCILLATOR 0' module interface. At the top, there are five input labels: TRIG, ROOT, INT1, INT2, and XPOSE. Below these are five horizontal sliders. The first slider is labeled 'ROOT' and has a value of -12. The second slider is labeled 'INT1' and has a value of 0. The third slider is labeled 'INT2' and has a value of 0. The fourth slider is labeled 'DTUNE' and has a value of 0. The fifth slider is labeled 'ATTACK' and has a value of 0. The sixth slider is labeled 'RELEASE' and has a value of 0. Below the sliders, there is a 'SCALE' dropdown menu set to 'Ionian'. To the right of the 'SCALE' dropdown is a 'WAVESELECTOR' dropdown menu set to 'SAW'. Below the 'WAVESELECTOR' dropdown is a 'XPOSE' dropdown menu set to 'SAW'. At the bottom, there are three output labels: TRIGGERED, CONSTANT, and EG.

PD-303

The PD-303 is an acid synthesizer in the style of the TB-303. It has a 16 step sequencer with SLIDE and ACCENT for each step. In addition to producing either a squarewave or a sawtooth, it outputs the pitch CV and envelope shape so that you can sequence anything else in the system. The PD-303 can store and recall 4 sequences. Click any of the four WRITE slots to store in slot 1-4 and click any of the LOAD slots to recall a stored sequence. Storing sequences with different SCALE and TRANSPOSE sequences can create some quite complex variations. You can step through the stored patterns with a trigger/gate signal at the LOAD(1-4) input.

CONTROLS:

STEPS1-16 - individual sliders to set the voltage level of each step

TRANSPOSE - transposes the sequence down 2octaves or up two octaves

CUTOFF - controls the cutoff frequency of the filter

RES - controls the resonance of the filter

ENVMOD - controls the amount of envelope sent to modulate the filter

ACCENT - controls the accent level

DECAY - controls the decay time of the accent envelope and the filter envelope

SLIDE - controls the overall portamento time for the SLIDE

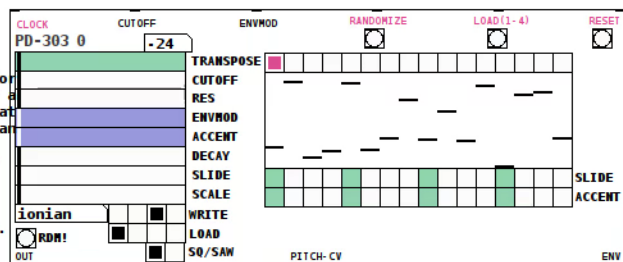
SCALE - choose scale for quantization

RDM! - button to instantly create a random pattern

WRITE - four slots of pattern storage

LOAD - four slots of pattern recall

SQ/SAW - select waveform (square or saw)



INPUTS:

CLK - trigger/gate/clock input

CUTOFF - cv input to modulate the CUTOFF

ENVMOD - cv input to modulate the ENVMOD amount

RANDOMIZE - trigger/gate input to randomize the entire pattern

LOAD(1-4) - trigger/gate input to step through stored pattern 1-4 sequentially

RESET - trigger/gate input to reset the sequencer

OUTPUTS:

OUT - synthesizer output

PITCH-CV - outputs the pitch CV of the sequencer (including SLIDES)

ENV - outputs the envelopes, including ACCENT

polysynth

POLYSYNTH:

The POLYSYNTH module is a 4 voice polyphonic synth voice. It needs a trigger/gate signal at the TRIG input to set off the internal envelope. Pitch CV is received at the PITCH. To have it play chords instead of single overlapping notes, use the TRIGGER-TRAIN module. (check the TRIGGER-TRAIN help-file for example)

CONTROLS:

PITCH - select the pitch offset of the synth

SAW->PULSE - crossfade from sawtooth to pulsewave

DTUNE - detunes the 4 voices slightly

ATTACK - attack time of the internal envelope

HOLD - hold time of the internal envelope, before entering release phase

RELEASE - release time of the internal envelope, before entering release phase

INPUTS:

TRIG - trigger the envelope

PITCH - cv input for the PITCH

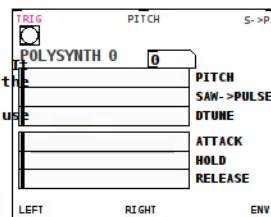
S->P - cv input for the SAW->PULSE control

OUTPUTS:

LEFT - outputs the left audio

RIGHT - outputs the right audio

ENV - outputs the envelope to use as a modulation source elsewhere in the system. For example, to modulate the cutoff of a filter placed after the POLYSYNTH.



polyfm

POLYFM:

The POLYFM module is a 4 voice polyphonic phase modulation synth voice. It needs a trigger/gate signal at the TRIG input to set off the internal envelope. There are two envelopes, one for the Carrier wave and one for the Modulator. Pitch CV is received at the PITCH. To have it play chords instead of single overlapping notes, use the TRIGGER-TRAIN module. (check the TRIGGER-TRAIN help-file for example)

CONTROLS:

PITCH - select the pitch offset of the synth

INDEX - control modulation index

RATIO - set the frequency ratio of the modulator

ATTACK - attack time of the internal envelope

HOLD - hold time of the internal envelope, before entering release phase

RELEASE - release time of the internal envelope, before entering release phase

INPUTS:

TRIG - trigger the envelope

PITCH - cv input for the PITCH

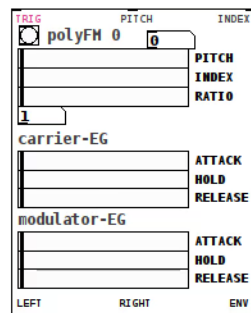
INDEX - cv input for index control

OUTPUTS:

LEFT - outputs the left audio

RIGHT - outputs the right audio

ENV - outputs the envelope (carrier-EG) to use as a modulation source elsewhere in the system. For example, to modulate the cutoff of a filter placed after the POLYSYNTH.



kick_drum_synthesizer

KICK-DRUM-SYNTHESIZER:

The "KICK" module is a synthesized kick drum.

CONTROLS:

PITCH - controls the pitch of kick drum

DECAY - controls the decay length of the envelope

EG2 - controls the decay length for the pitch envelope

SWEEP - controls the amount of EG2 sent to modulate pitch

WFORM - controls a three-way-crossfader between sine, saw and square

INPUTS:

TRIG - trigger/gate input to trigger the sound

PITCH - input for cv-control over pitch

DECAY - input for cv-control over decay

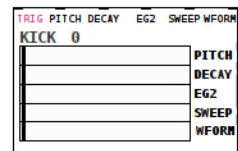
EG2- input for cv-control over eg2

SWEEP- input for cv-control over sweep

WFORM- input for cv-control over WFORM

OUTPUTS:

OUT - output



snare_drum_synthesizer

SNARE-DRUM-SYNTHESIZER:

The "snare" module is a simple synthesized snare drum module. It produces snare drum sounds in the vein of TR-808.

CONTROLS:

PITCH - controls the pitch of snare drum

DECAY -controls the decay length of the pitch

NOISE - controls the decay length of the noise

COLOR - filtering of the noise for different colors

INPUTS:

TRIG - trigger/gate input to trigger the sound

PITCH - input for cv-control over pitch

DECAY - input for cv-control over decay

NOISE - input for cv-control over noise

COLOR - input for cv-control over color

OUTPUTS:

OUT - output

TRIG	PITCH	DECAY	NOISE	COLOR
SNARE 0				
				PITCH
				DECAY
				NOISE
				COLOR

noise

NOISE-SOURCE

The **NOISE** module produces a constant white noise at the output.

OUTPUTS:

OUT - output

NOISE

looper

LOOPER:

The "LOOPER" module can record signals from your patch and loop them. It works like a live sampler. Press the RECORD toggle to record and stop recording. The recording can be automated by trigger signals at the REC input. The LOOPER can record CV signals as well as audio. It keeps the last recorded loop in memory when you quit PD, however when you launch the patch again it needs 5-7 seconds to load the recording.

CONTROLS:

SPEED - control the playback speed of the recorded loop

RECORDS - toggle recording on/off

ORIGINAL - press the button to get back to the original speed of the recording

INPUTS:

IN-L - left input

IN-R - right input

SPEED - cv input for modulating the SPEED parameter

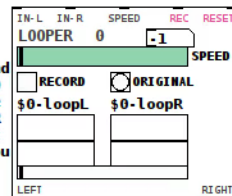
REC - trigger input to start/stop recording

RESET - trigger input to start the recording from the beginning

OUTPUTS:

LEFT - sample left channel output

RIGHT - sample right channel output

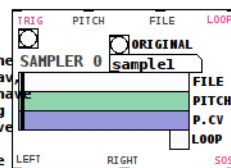


sampler

SAMPLER:

The "SAMPLER" module can load 8 different .wav tracks. The files are located in ../samples/ and are named sample1.wav, sample2.wav, sample3.wav etc. To change the samples you have to replace those files with your own files, while keeping the same name structure. This might seem counter-intuitive but the advantage of this system is that your files will always be pointed to the right location and you can move your project main folder anywhere and to different platforms without losing your samples. If you want a sampler that can load files from anywhere on your harddrive I suggest you try the excellent sampler module made by Claude Barker. More info about that at:

<https://github.com/megalon/pd-AUTOMATONISH-sampler>



CONTROLS:

FILE - select between sample 1-8

PITCH - change the pitch of the sample

P.CV - attenuator for cv at the PITCH input

LOOP/ISHOT - toggle between looping mode and lshot mode. In lshot mode the sampler needs a trigger signal at the TRIG input to fire the sample.

INPUTS:

TRIG - trigger input to play sample in lSHOT mode

PITCH - cv input for PITCH

FILE - cv input for modulating what sample is active

LOOP - trigger input to switch between LOOP and lSHOT mode

OUTPUTS:

LEFT - sample left channel output

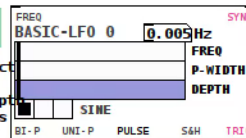
RIGHT - sample right output

SOS - trigger output at the start of each sample

basic_lfo

BASIC-LF0:

The "BASIC-LF0" module is low frequency oscillator. Select between sine, triangle or sawtooth wave and a separate pulsewave output. There is control over frequency and depth. The BASIC-LF0 is an excellent clock/trigger module and as long as DEPTH is set to maximum it will trigger any pink input in the system with the BI-P, UNI-P or PULSE output. The TRIG output is independent of DEPTH and will always trigger at the start of each cycle. An additional sample and hold unit gives a random stepped voltage each cycle.



CONTROLS:

FREQ - controls the pitch of the lfo

WAVEFORM SELECTOR - choose between sine, triangle or sawtooth

P-WIDTH - controls the pulsewidth of the PULSE output.

DEPTH - control over lfo depth

INPUTS:

FREQ - input for controlling the lfo-frequency with CV

SYNC - trigger/gate input for lfo-sync

OUTPUTS:

BI-POLAR - outputs a bi-polar sine, saw or triangle

UNI-POLAR - outputs a uni-polar sine, saw or triangle

PULSE - outputs a uni-polar pulsewave

S&H - outputs an held random voltage each cycle

TRIG - outputs short trigger each cycle--independently of the DEPTH control

sinebank

SINEBANK:

The "SINEBANK" consists of four sinewave oscillators. By toggling the LF0 toggle the range can be switched between audiorate or low frequency rate. Individual outputs as well as a SUM out is provided.

CONTROLS:

1 - controls the pitch of sinewave 1

2 - controls the pitch of sinewave 2

3 - controls the pitch of sinewave 3

4 - controls the pitch of sinewave 4

LF0 - toggle between audiorate and LF0

INPUTS:

CV1 - input for controlling sinewave 1

CV2 - input for controlling sinewave 2

CV3 - input for controlling sinewave 3

CV4 - input for controlling sinewave 4

OUTPUTS:

S1 - sinewave 1 output

S2 - sinewave 2 output

S3 - sinewave 3 output

S4 - sinewave 4 output

SUM - mix of all four sinewaves

CV1 CV2 CV3 CV4

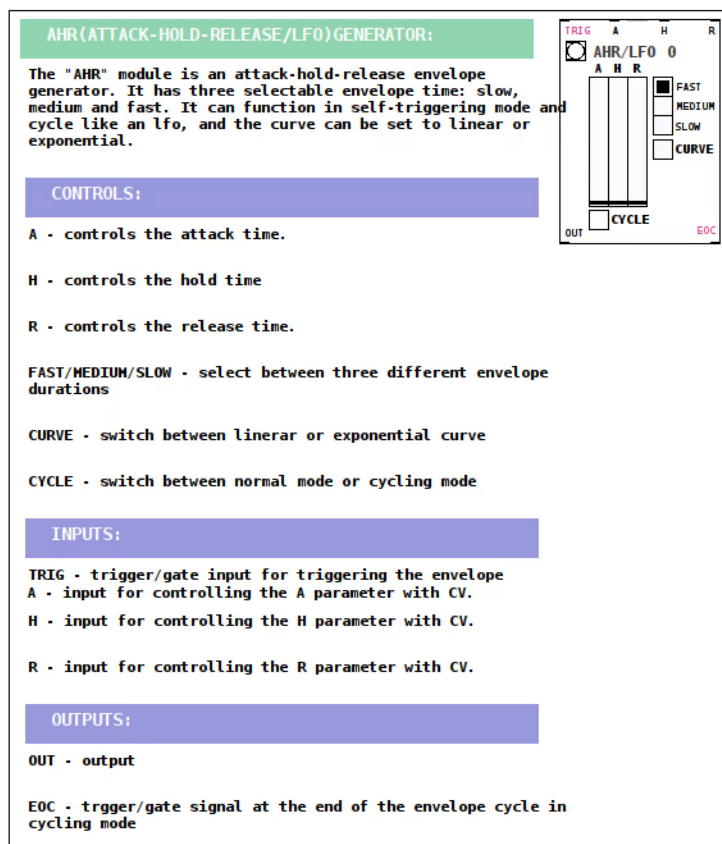
SINEBANK 0

1 2 3 4

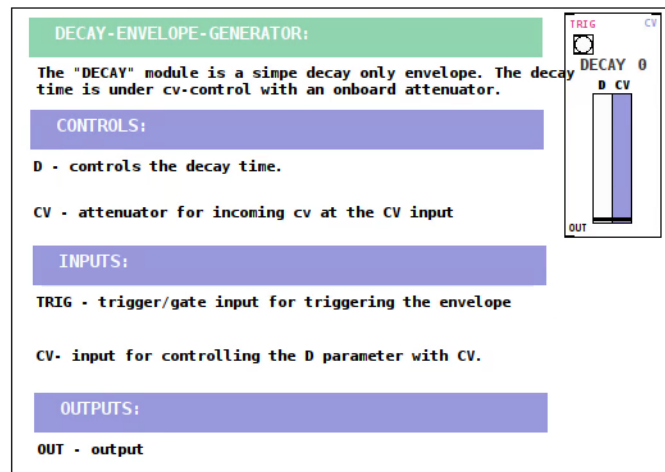
LF0

S1 S2 S3 S4 SUM

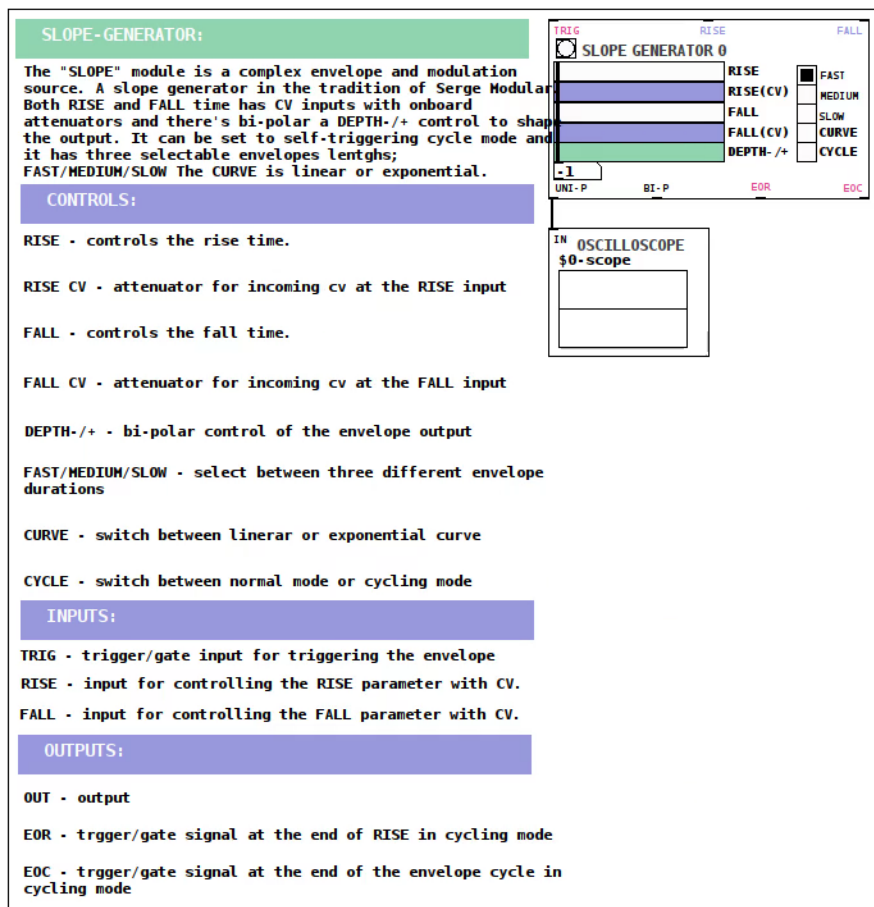
attack_hold_release_lfo_generator



decay_envelope_generator



slope_generator



adsr

ADSR:

The "adsr" module is an attack-decay-sustain-release envelope best used with gate signals. AUTOMATONISM is mainly built around trigger signals. However, the GATE-SEQ, or variable pulsewave lfos are valid companions to the ADSR. Also, use of external controllers or keyboards could be used with the module.

CONTROLS:

ATTACK - controls the attack time.

DECAY - controls the decay time

SUSTAIN - controls the sustain level

RELEASE - controls the release time

CURVE - croosfade bewteen linear and exponential curve

INPUTS:

GATE - gate input for triggering the ADRS

OUTPUTS:

OUT - output

GATE

ADSR 0

ATTACK

DECAY

SUSTAIN

RELEASE


CURVE

random_gates

RANDOM-GATES:	
<p>The "RANDOM-GATES" module generates random trigger/gate signals by feeding random tempi to a metronome. You have control over the rhythmic intensity as well as the probability of a trigger/gate happening.</p>	
<p>CONTROLS:</p> <p>INTENSITY - controls the intensity of random gates</p> <p>PROB. - controls the probability of a gate happening</p>	
<p>OUTPUTS:</p> <p>OUT - output</p>	

RANDOM GATES 0

	INTENSITY
	PROB.

 OUT

random_voltages

RANDOM-VOLTAGES

The "RANDOM-VOLTAGES" module is a random voltages generator. When a trigger/gate signal is received at the **CLOCK-IN** input it will produce a random voltage. Range can be set with the **RANDOM** slider. In addition you can add a slew to the voltage with the **SLEW** slider. The module outputs both a **BI-POLAR** voltage and a **UNI-POLAR** voltage.

CONTROLS:

RANDOM - attenuates the range of randomness

SLEW - applies a slew to the output

INPUTS:

CLOCK-IN - trigger/gate input

OUTPUTS:

BI-P - bi-polar random voltage

UNI-P - uni-polar random voltage

CLOCK-IN

☐ **RANDOM-VOLTAGES** 0

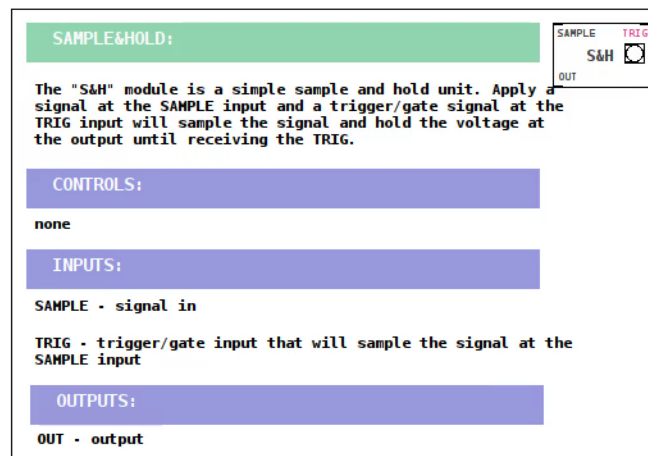
RANDOM

SLEW

BI-P

UNI-P

sample_&_hold



8_steps_sequencer

8STEPS

The "8STEPS" module is a simple eight step sequencer with onboard quantization. 8steps requires a trigger/gate/clock signal at the CLOCK-IN input to step forward. There are two cv inputs. CV(PRE-Q) modulates the sequencer before the quantizer so the sequencer will still be in tune when applying modulation here. CV(POST-Q) adds the CV after the quantizer so any CV at the input will transpose the sequence.

CONTROLS:

STEPS1-8 - individual sliders to set the voltage level of each step

SCALE - choose scale for quantization

INPUTS:

CLOCK-IN - trigger/gate/clock input

CV(PRE-Q) - modulation input. CV added to the sequence before quantization.

CV(POST-Q) - modulation input. CV added to the sequence after quantization.

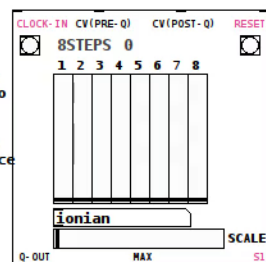
RESET - trigger/gate input to force reset the sequencer to step 1

OUTPUTS:

Q-OUT - quantized output

OUT2 - un-quantized output that outputs the full voltage range. This output is often more useful if you're using the sequencer to modulate other parameters than pitch of an oscillator.

S1 - trigger/gate signal at start of sequence loop



mega_sequencer

MEGA-SEQUENCER

The "MEGA-SEQUENCER" is an advanced sequencer tool. It has 16 steps and a randomization function per step. Ticking the toggle box beneath each step will randomize the value for that step. There are controls for choosing scale (SCALE) and transposing the sequence - 24 semitones to +24 semitones (XPOSE). There is a bang button called RDM-PATTERN that will supply new values randomly to all the steps. The button RESET-TOGGLES resets all random-step toggles to zero. The MEGA-SEQUENCER can store and recall 4 sequences. Click any of the four WRITE slots to store in slot 1-4 and click any of the LOAD slots to recall a stored sequence. Storing sequences with different SCALE and XPOSE sequences can create some quite complex variations. You can step through the stored patterns with a trigger/gate signal at the LOAD(1-4) input.

Q-OUT

OUT2

CONTROLS:

STEPS1-16 - individual sliders to set the voltage level of each step

SCALE - choose scale for quantization

XPOSE - transpose the sequence -24 semitones to +24 semitones.

RANDOM-STEP - 16 toggleboxex underneath each step which when activated will force a random voltage on that particular step.

RDM-PATTERN - button to instantly create a random pattern

RESET-TOGGLES - button to reset all RANDOM-STEP toggles to zero.

WRITE - four slots of pattern storage

LOAD - four slots of pattern recall

INPUTS:

CLK - trigger/gate/clock input

RANDOMIZE - trigger/gate input to randomize the entire pattern

LOAD(1-4) - trigger/gate input to step through stored pattern1-4 sequentially

OUTPUTS:

Q-OUT - quantized output

OUT2 - un-quantized output that outputs the full voltage range. This output is often more useful if you're using the sequencer to modulate other parameters than pitch of an oscillator.

trigger_sequencer

TRIGGER-SEQ:

The "TRIGGER-SEQ" is a trigger/gate sequencer. It has a maximum of 16 steps but can be set to any number of steps. The TRIGGER-SEQ can store and recall 4 sequences. Click any of the four WRITE slots to store in slot 1-4 and click any of the LOAD slots to recall a stored sequence. You can step through the stored patterns with a trigger/gate signal at the LOAD(1-4) input.

CLOCK-IN	LOAD(1-4)	RESET													
TRIGGER SEQ 0															
<div><div></div><div></div><div></div><div></div><div></div><div></div><div></div><div></div><div></div><div></div><div></div><div></div><div></div><div></div><div></div><div></div></div>															
GATE-ON/OFF				STORE				LOAD							
<div><div></div></div>				<div><div></div><div></div><div></div><div></div></div>				<div><div></div><div></div><div></div><div></div></div>							

CONTROLS:

STEPS1-16 - individual toggle boxes to set the step on or off.

WRITE - four slots of pattern storage

LOAD - four slots of pattern recall

INPUTS:

CLOCK-IN - trigger/gate/clock input

LOAD(1-4) - trigger/gate input to step through stored pattern1-4 sequentially

RESET - trigger/gate input to reset the sequencer

OUTPUTS:

OUT - output

gate_sequencer

GATE-SEQ:

The "GATE-SEQ" is an eight-step gate sequencer with adjustable gate-length and on/off per step. All parameters can be randomized by clicking the RDM! button manually or by CV.

CONTROLS:

STEPS1-8 - individual toggle boxes to set the step on or off.

GATE-LENGTH SLIDERS: - set the gate length per step

INPUTS:

CLOCK - trigger/gate/clock input

RESET - trigger/gate input to reset the sequencer

RDM! - trigger/gate input to randomize all parameters

OUTPUTS:

OUT - output

CLOCK RESET RDM!

GATE-SEQ 0

<input checked="" type="checkbox"/>	<input type="checkbox"/>	<input type="checkbox"/>	<input type="checkbox"/>	<input type="checkbox"/>	<input type="checkbox"/>	<input type="checkbox"/>	<input type="checkbox"/>

BI-P UNI-P

lowpass_filter

LOWPASS-FILTER

The "LP-FILTER" module is a resonant lowpass filter with control over cutoff frequency and resonance.

CONTROLS:

CUTOFF - controls the cutoff frequency of the filter.

Q - controls the resonance

FM-/+ - bi-polar attenuator for incoming CV at the CUTOFF input

INPUTS:

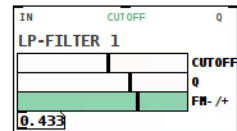
IN - signal input

CUTOFF - input for cv-control over the CUTOFF with onboard bi-polar attenuator

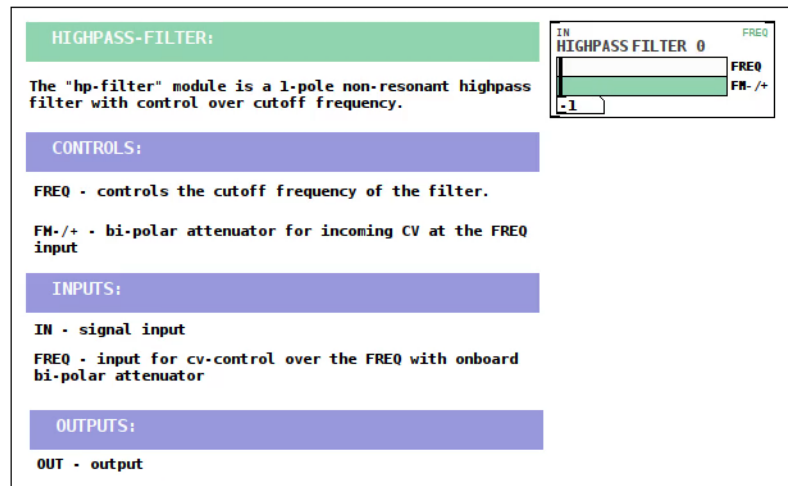
Q - input for cv-control over the Q.

OUTPUTS:

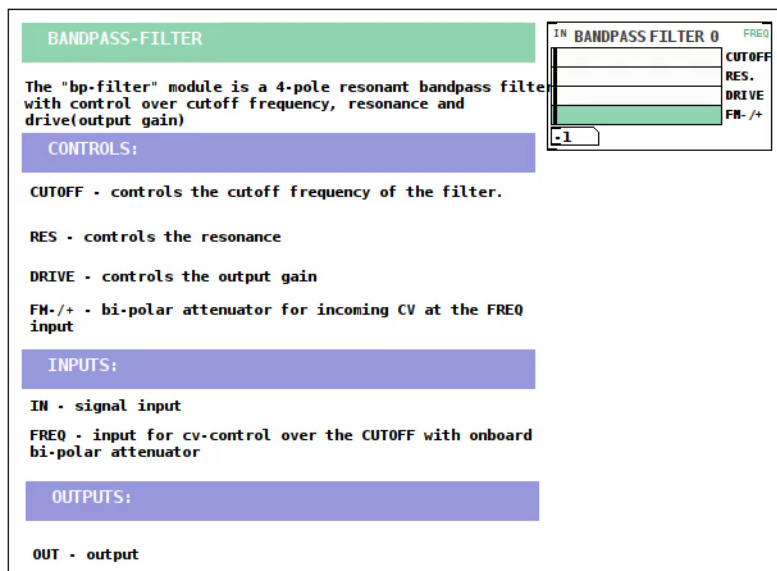
OUT - output



highpass_filter



bandpass_filter



ladder_filter

LADDER-FILTER

The "LADDER-FILTER" module is a resonant lowpass filter with cv-control over cutoff frequency. It has slightly more character than LP-FILTER with more warmth and a steeper slope.

CONTROLS:

FREQ - controls the cutoff frequency of the filter.

Q - controls the resonance

FM-/ + - bi-polar attenuator for incoming CV at the CUTOFF input

INPUTS:

IN - signal input

FREQ - input for cv-control over the CUTOFF with onboard bi-polar attenuator

OUTPUTS:

OUT - output

IN

LADDER-FILTER

0

FREQ

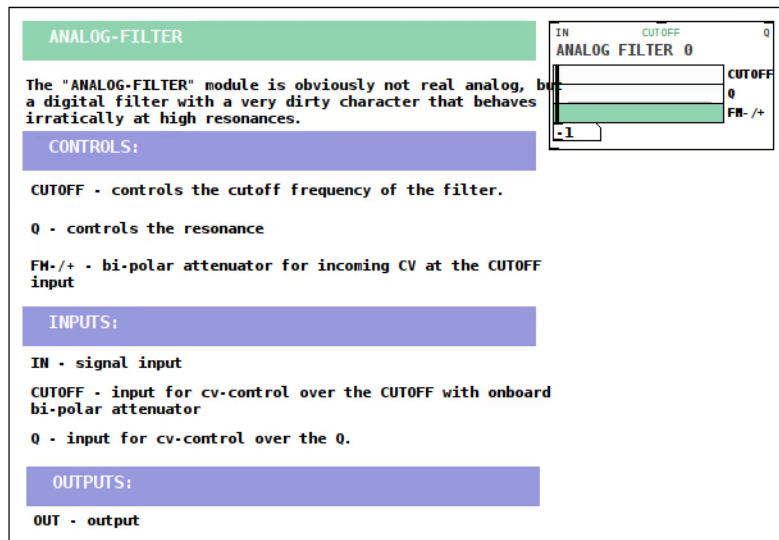
0

Q

FM-/ +

-1

analog_filter



formant_filter

FORMANT-FILTER:

The "FORMANT" module consists of multiple bandpass filters set to frequencies that mimics human vowel sounds. FORMANT works best with signals high in harmonic content.

CONTROLS:

VOWELS - scan through different vowels.

Q - control over Q (resonance)

INPUTS:

IN - signal input

VOWELS - input for cv-control over the FORMANTS.

OUTPUTS:

OUT - output

IN

FORMANT FILTER 0

VOWELS

Q

bitcrush

BITCRUSH

The "BITCRUSH" module is a crude digital distortion unit.

CONTROLS:

DRY - controls amount of unaffected dry signal at the output

CRUSH - controls the amount of bitcrush to the signal

INPUTS:

IN - signal input

CRUSH - input for cv-control over the CRUSH parameter.

DRY - input for cv-control over the DRY parameter.

OUTPUTS:

OUT - output

IN

CRUSH

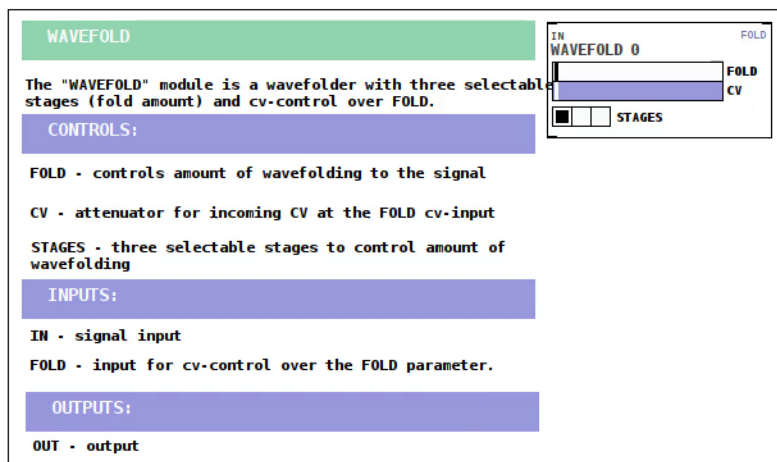
DRY

BITCRUSH 0

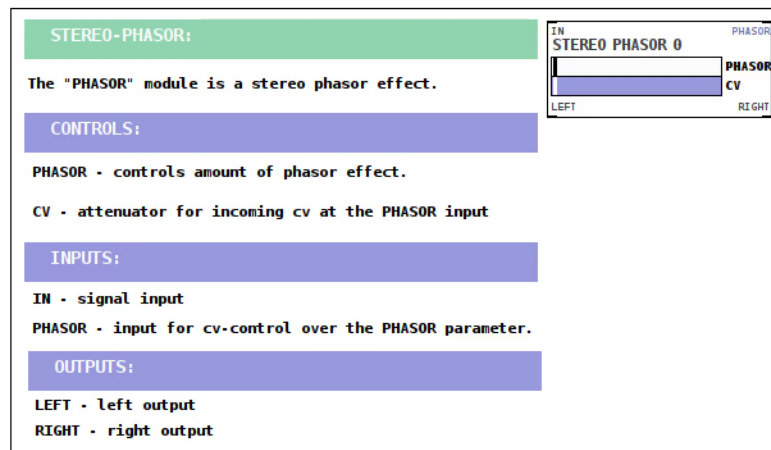
DRY

CRUSH

wavefold



stereo_phasor



combfilter

COMBFILTER:

The "COMBFILTER" module is delay fx unit. Delay times range from 1ms - 100ms.

CONTROLS:

DELAY - controls delay time from 1ms - 100ms

FEEDBACK - controls amount of feedback

LOWPASS - set lowpass filter frequency cutoff at the output

HIGHPASS - set highpass filter frequency cutoff at the the output

DRIVE - set drive (gain) at the output

DRY/WET - control the balance between dry and wet signal

INPUTS:

IN - signal in

DELAY -input for cv-control over the DELAY parameter

OUTPUTS:

OUT - output

IN

COMBFILTER 0

DELAY

	DELAY
	FEEDBACK
	LOWPASS
	HIGHPASS
	DRIVE
	DRY/WET

granular_delay

GRANULAR-DELAY:

The "GRANULAR-DELAY" module is an experimental granular delay fx unit.

CONTROLS:

PITCH - change the pitch of the signal

GRAINSIZE - controls the size of the grains/delay

REVERB - sets reverb level

DRY/WET - control the balance between dry and wet signal

INPUTS:

IN - signal in

PITCH -input for cv-control over the PITCH parameter

GRAINSIZE -input for cv-control over the GRAINSIZE parameter

REVERB -input for cv-control over the FBACK parameter

D/W -input for cv-control over the DRY/WET parameter

OUTPUTS:

OUT - output

IN	PITCH	GRAIN	REVERB	D/W
GRANULAR DELAY 0			-1	
			PITCH	
			GRAIN	
			REVERB	
			DRY/WET	

millerverb

MILLERVERB:

The "MILLERVERB" module is a stereo reverb unit based on the reverberation example code from Miller Puckette's book.

CONTROLS:

REVERB - controls amount of reverb/size of the room

LOWPASS - controls the cutoff frequency of a non resonant lowpass filter

DRY/WET - controls the balance between dry and wet signal

GAIN - controls the output gain

INPUTS:

IN - signal in

REVERB -input for cv-control over the REVERB parameter

LPASS -input for cv-control over the LOWPASS parameter

OUTPUTS:

LEFT - left output

RIGHT - right output

IN

REVERB

LPASS

MILLERVERB 0

REVERB

LOWPASS

DRY/WET

GAIN

LEFT

RIGHT

stereo_delay

STEREO-DELAY:

The "STEREO-DELAY" module is a mono input - stereo output delay with control over delay rate, feedback and amount of lowpass filtering.

CONTROLS:

RATE-LEFT - controls delay rate of the left channel. Delay times from 10ms - 500ms

RATE-RIGHT - controls delay rate of the right channel. Delay times from 10ms - 500ms

FEEDBACK - controls amount of feedback

LOWPASS - set lowpass filter frequency cutoff at the output

DRY/WET - control the balance between dry and wet signal

INPUTS:

IN - signal in

RATE-L - input for cv-control of the RATE-LEFT parameter

RATE-R - input for cv-control of the RATE-RIGHT parameter

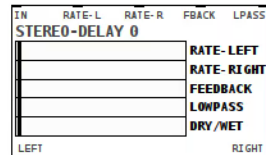
FBACK - input for cv-control of the FEEDBACK parameter

LPASS - input for cv-control of the LOWPASS parameter

OUTPUTS:

LEFT - left output

RIGHT - right output



chorus

CHORUS:

The "chorus" module is a mono input to stereo output chorus effect.

CONTROLS:

RATE - controls the chorus rate

DEPTH - controls the depth of chorus effect

INPUTS:

IN - signal in

RATE - input for cv-control of the RATE parameter

DEPTH - input for cv-control of the DEPTH parameter

OUTPUTS:

LEFT - left output

RIGHT - right output

IN

CHORUS 0

LEFT

RATE

DEPTH

RIGHT

overdrive

OVERDRIVE:

The "overdrive" module is distortion/overdrive/fuzz effect.

CONTROLS:

DRIVE - controls the distortion/overdrive/fuzz amount

TONE - controls the EQ of the fuzz

INPUTS:

IN - signal in

DRIVE - input for cv-control of the DRIVE parameter

TONE - input for cv-control of the TONE parameter

OUTPUTS:

OUT - output

IN	DRIVE	TONE
OVERDRIVE	0	
		DRIVE
		TONE

graphic_eq

GRAPHIC-EQ:

The "GRAPHIC-EQ" module is stereo 7-band equalizer. Use it to shape the tone of your sound sources.

CONTROLS:

- 100 - set the level of the 100Hz bandpass filter
- 200 - set the level of the 200Hz bandpass filter
- 400 - set the level of the 400Hz bandpass filter
- 800 - set the level of the 800Hz bandpass filter
- 1600 - set the level of the 1600Hz bandpass filter
- 3200 - set the level of the 3200Hz bandpass filter
- 6400 - set the level of the 6400Hz bandpass filter

CV(LOW) - attenuator for CV at the LOW input

CV(MID) - attenuator for CV at the MID input

CV(HIGH) - attenuator for CV at the HIGH input

Q - global control for the Q of all bandpass filters

LEVEL - adjust output level of the GRAPHIC-EQ

INPUTS:

LEFT - left audio input

RIGHT - right audio input

LOW - cv input for the purple bandpass filters

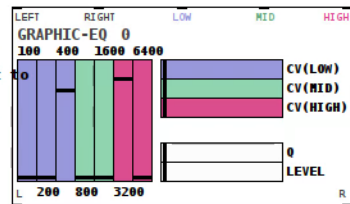
MID - cv input for the green bandpass filters

HIGH - cv input for the pink bandpass filters

OUTPUTS:

L - left channel output

R - right channel output



0170_megaverb

MILLERVERB:

The "MEGAVERB" module is a reverb unit with larger spaces than the MILLERVERBm module. The code is an adaptation of the [rev2~] object into Automatonism signal flow.

CONTROLS:

SIZE - controls size of the room

DAMPING - controls the dampening of the room

LOWPASS - controls the cutoff frequency of a non resonant lowpass filter at the output

DRY/WET - controls the balance between dry and wet signal

INPUTS:

IN - signal in

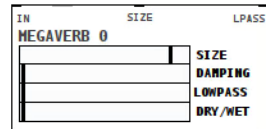
SIZE -input for cv-control over the SIZE parameter

LPASS -input for cv-control over the LOWPASS parameter

OUTPUTS:

LEFT - left output

RIGHT - right output



pitchshifter

PITCHSHIFTER:

The PITCHSHIFTER module is an adaptation of the code in the Pure Data examples (G09.pitchshift.pd) for the Automatonism signal flow. It takes a signal and shifts the pitch with a range of down two octaves to up two octaves.

CONTROLS:

PITCH-/+ - shift the pitch of the input signal

CV - attenuator for incoming CV at the PITCH-/+ cv-input

DTY/WET - dry/wet control

INPUTS:

IN - signal input

CV - input for cv-control over the PITCH-/+ parameter.

OUTPUTS:

OUT - output

IN

PITCHSHIFTER 0 -24

CV

PITCH-/+

CV

DRY/WET

vca

VCA:

The "VCA" module is a voltage controlled amplifier/attenuator. It is one of the most useful modules in any modular system. Typically a vca is most commonly used to dynamically change a signal's amplitude over time in tandem with an envelope generator. But the same could be applied to any signal that you want to change dynamically over time in a patch. Basic example: instead of patching an LFO directly to the filter cutoff, go through a vca and control that vca with another LFO. This is called "modulate the modulator" and can really give some life to your patches.

CONTROLS:

VCA - controls the level of the signal present at the **IN** input.

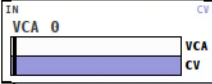
CV - attenuator for incoming cv at the CV input

INPUTS:

IN - signal in

INPUTS:

OUT - output



lpg_lowpass_gate

LPG(LOWPASS-GATE):

The "LPG" module is a lowpass gate; a combination of vca and lowpass filter. A short trigger/gate signal at the CV input is enough to open the gate and you can set the response with the RING/DAMP control. A lowpass gate is especially good for percussive "bongo" type sounds but can be used just like you would use a vca or a lowpass filter as well.

CONTROLS:

LEVEL - controls the level of the signal input. The response of the slider is determined by the RING/DAMP control.

RING/DAMP - controls how long you want the lpg to ring after excitation.

CV - attenuator for incoming CV at the CV input

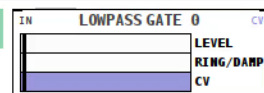
INPUTS:

IN - signal in

CV - input for cv-control of the LEVEL parameter

OUTPUTS:

OUT - output



xfade

XFADEx:

The "xfade module is a simple crossfader. Two signal inputs can be crossfaded with a slider manually or with cv.

CONTROLS:

XFADEx - crossfades between signal-in-1 and signal-in-2

INPUTS:

IN-1 - signal in 1
IN-2 - signal in 2
XFADEx - input for cv-control of the XFADEx parameter

OUTPUTS:

OUT - output

IN-1IN-2XFADEx

XFADEx 0

OUTINV. OUT

3way_xfade

3WAY-XFADE:

The "3WAY-XFADE" module is a three-way crossfader. Three signal inputs can be crossfaded with a slider manually or with cv.

CONTROLS:

XFADE - crossfades between signal 1, 2 and 3

INPUTS:

IN-1 - signal in 1
IN-2 - signal in 2
XFADE - input for cv-control of the XFADE parameter

OUTPUTS:

OUT - output

123XFADE

3WAY-XFADE 0

XFADE

polarizer

POLARIZER:

The "polarizer" module is a simple polarizer/attenuverter/offset control for finetuning CV sources. Any signal can be scaled down, inverted into the negative or offsetted.

CONTROLS:

POL-/+ - attenuvert the signal

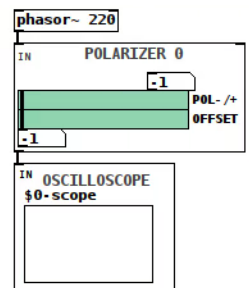
OFFSET - offset the signal with DC

INPUTS:

IN - signal in

OUTPUTS:

OUT - output



EXAMPLE: move the POL-/+ or OFFSET slider and watch the scope

mixer2

MIXER2:

The "MIXER2" module is a dual channel summing mixer. It can be used with any signal source; audio, cv or trigger/gate.

CONTROLS:

CHANNEL1 - controls the level of channel 1

CHANNEL2 - controls the level of channel 2

INPUTS:

CH1 - signal in

CH2 - signal in

OUTPUTS:

OUT - output

CH1

MIXER 0

CH2

CH1

CH2

OUT

UNITY

mixer4

MIXER4:

The "mixer4" module is a four-channel summing mixer. It can be used with any signal source; audio, cv or trigger/gate.

CONTROLS:

CHANNEL1 - controls the level of channel 1

CHANNEL2 - controls the level of channel 2

CHANNEL3 - controls the level of channel 3

CHANNEL4 - controls the level of channel 4

INPUTS:

CH1 - signal in

CH2 - signal in

CH3 - signal in

CH4 - signal in

OUTPUTS:

OUT - output

CH1	CH2	CH3	CH4
MIXER 0			
			CH1
			CH2
			CH3
			CH4
OUT			UNITY

toggle_matrix

TOGGLE-MATRIX:

The "TOGGLE-MATRIX" module is a classic matrix mixer with toggle switches instead of potentiometers. Four signal inputs can be turned on/off in four different ways at the A-D outputs.

CONTROLS:

TOGGLES - 1A-4A
TOGGLES - 1B-4B
TOGGLES - 1C-4C
TOGGLES - 1D-4D

INPUTS:

1 - input one
2 - input two
3 - input three
4 - input four

OUTPUTS:

A - output A
B - output B
C - output C
D - output D

1234

TOGGLE-MATRIX 0

1				
2				
3				
4				
	A	B	C	D
A				
B				
C				
D				

maestro4

MAESTR04:

The "MAESTR04" module is a digital-to-analog converter and four channel panning mixer. It is a module that will pass on sound to your speakers. It has manual and cv-control over panning.

CONTROLS:

CH1, CH2, CH3 & CH4 - controls gain of equivalent channel

PAN - control panning of equivalent channel

MASTER - controls the volume of the sum of channels 1-4

INPUTS:

CH1-4 - channel inputs

PAN1-4 - cv input for panning

OUTPUTS:

none/

CH1 PAN1 CH2 PAN2 CH3 PAN3 CH4 PAN4

MAESTR04 0

CH1 CH2 CH3 CH4 MASTER

PAN1 PAN2 PAN3 PAN4

CH1 CH2 CH3 CH4 LEFT RIGHT

clock

CLOCK:

The "CLOCK" module is a simple clock/trigger/gate generator. Control the clock-rate or turn the clock on/off. Notice that there are other ways to generate clocks in the patch editor. For example using BASIC-LFO or the EOC of a cycling envelope.

CONTROLS:

CLK_RATE - controls the clock-rate

ON/OFF - turn the clock on/off

INPUTS:

CLK_RATE - input for cv-control of the clock rate

ON/OFF - trigger/gate input for turning the clock on/off

OUTPUTS:

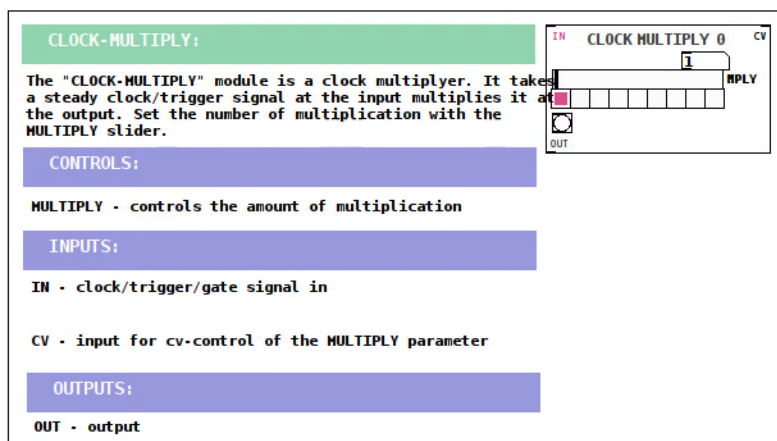
OUT - output

CLK-RATE CLOCK 0 ON/OFF

1970 RATE ON/OFF

OUT

clock_multiply



clock_divider_even

CLOCK-DIVIDER-EVEN

The "clock-divider-even" module is a simple clock divider with even subdivisions of the clock. Available divisions are /2, /4, /8 and /16.

CONTROLS:

none/

INPUTS:

IN - trigger/gate input to be divided

OUTPUTS:

/2 - input divided by 2

/4 - input divided by 4

/8 - input divided by 8

/16 - input divided by 16

IN

CLOCK DIVIDER

/2

/4

/8

/16

clock_divider_even

CLOCK-DIVIDER-ODD:

The "CLOCK-DIVIDER-ODD" module is a simple clock divider with odd subdivisions of the clock. Available divisions are /2, /5, /7 and /9.

CONTROLS:

none/

INPUTS:

IN - trigger/gate input to be divided

OUTPUTS:

/3 - input divided by 3
/5 - input divided by 5
/7 - input divided by 7
/9 - input divided by 9

IN

CLOCK DIVIDER

/3

/5

/7

/9

vc_counter

VC-COUNTER:

The "VC-COUNTER" module is voltage controlled counter/clock divider. Incoming trigger/gate/clock can be subdivided between 1-64. Useful module for setting a very specific code divisions not available immediately from the other clock-divider modules. You could also feed it CV to dynamically control the subdivisions.

CONTROLS:

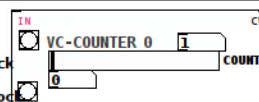
COUNT - set number of subdivisions

INPUTS:

IN - trigger/gate input to be divided

OUTPUTS:

OUT - divided output



trigger_delay

TRIGGER-DELAY:

The "trigger-delay" module is a simple trigger delay. Trigger/gate signal at the input goes to the output after a delay time in milliseconds set by the DELAY slider.

CONTROLS:

DELAY - controls delay time in ms

INPUTS:

IN - trigger/gate input

OUTPUTS:

OUT - delayed trigger/gate output

IN

TRIGGER DELAY 0

10

DELAY

OUT

trigger_train

TRIGGER-TRAIN:

The TRIGGER-TRAIN module takes one trigger/gate signal and outputs four in series. The time between the four triggers are set with the DELAY slider. It is particularly useful to make the POLYSYNTH and POLYFM play chords. See example to the right.

CONTROLS:

DELAY - controls delay time in between triggers

INPUTS:

IN - signal in

CV - input for cv-control of the DELAY parameter

OUTPUTS:

OUT - output

CLK-RATE

CLOCK 0

ON/OFF

RATE

1970

ON/OFF

OUT

IN

CV

TRIGGER-TRAIN 0

DELAY

20

TRIG

PITCH

S->P

POLYSYNTH 0

0

PITCH

SAW->PULSE

DTUNE

ATTACK

HOLD

RELEASE

LEFT

RIGHT

ENV

CLOCK-IN

CV(PRE-Q)

CV(POST-Q)

RESET

8STEPS 0

1

2

3

4

5

6

7

8

ionian

SCALE

Q-OUT

MAX

SL

voltage_inverter

VOLTAGE-INVERTER:

The "INVERTER" module is a voltage inverter. Invert the polarity of any signal. Mainly used for CV manipulation

CONTROLS:

none/

INPUTS:

IN - signal in

OUTPUTS:

OUT - output

IN

VOLTAGE
INVERTER

OUT

oscilloscope

OSCILLOSCOPE:

The "SCOPE" module is a simple oscilloscope. Use it frequently to gain a deeper understanding of signals!

CONTROLS:

none/

INPUTS:

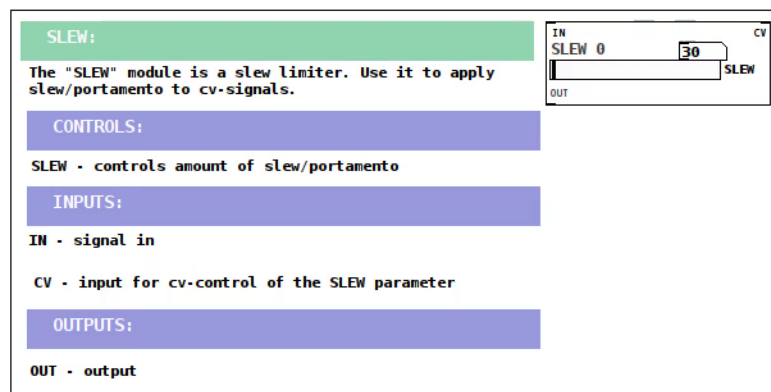
IN - signal in

OUTPUTS:

OUT - output

IN OSCILLOSCOPE
\$0-scope

slew



probability

PROBABILITY:

The "PROBABILITY" module is probability logic module for trigger/gate signals. It takes a signal at the input and passes it forward either to the OUT1 or OUT2 output. The PROB slider set the probability of which route the signal goes.

CONTROLS:

PROB - sets the probability of either signal passing through OUT1 or OUT2

INPUTS:

IN - signal in
CV - input for cv-control of the PROB paramater

OUTPUTS:

OUT1 - output1
OUT2 - output2

IN

PROBABILITY 0

CV

PROB

OUT1

OUT2

quantizer

QUANTIZER:

The "QUANTIZER" module is a simple quantizer. It takes any cv signal and quantizes the range into intervalic scales. Patch any signal to the IN input and get the quantized voltage at the output. Available scales: ionian, dorian, phrygian, lydian, aeolian, whole, chromatic, harmonic-major, lydian-minor, ancient, tetra, indian, pelog, slendro

CONTROLS:

SCALE - choose scale

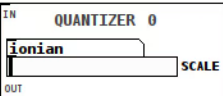
INPUTS:

IN - signal in

TRIG - trigger/gate input to sample the CV at the IN input and force quantization.

OUTPUTS:

OUT- output



4to1_switch

4to1-SWITCH:

The "4to1-SWITCH" is a sequential switch that takes 4 inputs and sequentially steps through the inputs at the output whenever a trigger/gate signal is received at the TRIG input.

CONTROLS:

/none

INPUTS:

1 - input 1
2 - input 2
3 - input 3
4 - input 4
TRIG - trigger/gate that triggers the sequential switch

OUTPUTS:

OUT - output

1234TRIG

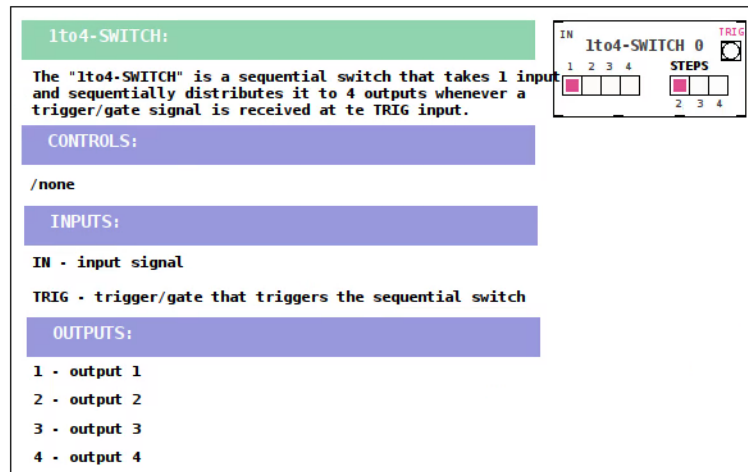
4to1-SWITCH 0

1234STEPS

1234

234

1to4_switch



env_follower

ENV-FOLLOWER:

The "ENV-FOLLOWER" is a simple envelope-follower. Use it to extract amplitude envelop shapes, gate signals and for frequency detection.

CONTROLS:

INPUT - Boost the input signal from 0 - *4

AMP-SMOOTH - smooth the amplitude shape

GATE-THRESH - select threshold needed to extract gate signal

AMPLITUDE-/+ - attenuueverter to shape the AMPLITUDE output

INPUTS:

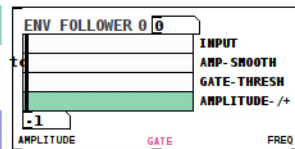
IN - signal in

OUTPUTS:

AMPLITUDE - envelope output

GATE - gate output

FREQ - frequency detection output



preset_manager

PRESET-MANAGER:

The "PRESET-MANAGER" module can store all parameters in a patch into 8 different slots. You can recall the saved slots manually, address them with CV or step through them sequentially with a trigger/gate signal. An additional MORPH control is available to set the time in milliseconds that it will take to reach the next parameter state. The preset-manager is a complex composition tool which allows you to experiment with different knob and parameter settings in a patch while being able to revert to a previous state. It is best used when a patch is finished and make sure to make a backup so you don't lose precious settings while experimenting.

CONTROLS:

STORE - First choose which slot to save to STORE(1-8)-- and then click the STORE button to save all current parameter values in a patch into the slot chosen by the STORE radio.

RECALL - immediately recall saved slots 1-8 by clicking the radio.

RECALL-CV - attenuator for incoming CV at the RECALL(CV) input.

MORPH - slider sets the morph time in milliseconds to glide to next value. Times from 0 seconds up to one minute.

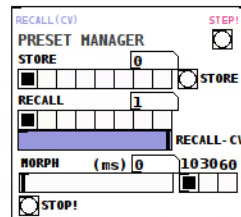
10 30 60 - determines maximum value of the MORPH slider. Set times between 0-10 seconds, 0-30 seconds, 0-60 seconds.

STOP! - click the STOP! button to stop a current morph between parameter values. It is useful to be able to cancel a morph if you need further editing and it can also be used creatively if the morph function reveals a new interesting parameter combination. If so, click STOP! and STORE.

INPUTS:

RECALL(CV) - input for CV-control over the RECALL parameter

STEP! -trigger/gate input that will step through presets sequentially 1-8



param_nudge

PARAM-NUDGE:

The "PARAM-NUDGE" module is an experimental tool made for mapping out interesting musical spaces. It allows you to nudge all parameters in a patch to search for new interesting sounds. Set how much you want the parameters to be moved with the +/- RANGE slider and click the NUDGE! button to nudge all parameters. PARAM-NUDGE is a good partner to the preset-manager module when trying to make interesting presets. (For practical reasons, the MAESTRO4 module is not affected by the PARAM-NUDGE module)

CONTROLS:

NUDGE - click button to nudge all parameters.

INPUTS:

NUDGE - trigger/gate input to nudge parameters

OUTPUTS:

/none

NUDGE
PARAM-NUDGE!

119

NUDGE!

-/+ RANGE

audio_recorder

AUDIO-RECORDER:

The AUDIO-RECORDER module takes a stereo input and records a 24bit .wav file to the folder called "recorded_wavs" in the project's main folder. It starts recording when you press the REC toggle and automatically labels your files as recording_1, recording_2, etc. Press the RST button to reset the counter, but previous recordings will be overdubbed. You can use the LEFT and RIGHT master outputs from the MAESTRO4 and patch them into the AUDIO-RECORDER. It is NOT recommended to use more than one AUDIO-RECORDER, it will likely cause PD and Automatonism to crash. Instead, use the MULTI-TRACK RECORDER MODULE to record several stems.

CONTROLS:

REC/STOP - Start and stop recording

RST - reset the recording number

LEVEL - set the overall gain of the incoming signals

INPUTS:

LEFT - left signal in

RIGHT - right signal in

OUTPUTS:

/none/

LEFT
AUDIO-RECORDER
REC/STOP 0 TAKE RST
\$0-left \$0-right
LEVEL

multitrack_recorder

MULTITRACK-RECORDER:

The MULTITRACK-RECORDER module takes maximum of 8 inputs and records 24bit .wav files to the folder `../recorder_wavs/multitrack/`

You can toggle channels on or off if you for example only want to record 4 stems. There is a LEVEL slider for the global recording volume. It starts recording when you press the REC toggle and automatically labels your files as "multitracker\$1_CH\$1_take\$1.wav". Press the RST button to reset the counter, but previous recordings will be overdubbed.

CONTROLS:

REC/STOP - Start and stop recording

RST - reset the recording number

LEVEL - set the overall gain of the incoming signals

CH1-8 ON/OFF TOGGLES - turn recording on or off for each channel

INPUTS:

CH1-CH8 - 8 channel inputs

OUTPUTS:

/none/

CH1CH2CH3CH4CH5CH6CH7CH8

MULTITRACK-RECORDER 0

REC/STOP 1

TAKE

RST

CH1

CH2

CH3

CH4

CH5

CH6

CH7

CH8

\$0-ch1

\$0-ch2

\$0-ch3

\$0-ch4

\$0-ch5

\$0-ch6

\$0-ch7

\$0-ch8

LEVEL

external_audio

EXTERNAL-AUDIO:

The "EXTERNAL-AUDIO" module is similar to the envelope follower moduler. It takes the audio input from your soundcard input channels 1 and 2, and outputs the audio, extracts amplitude envelop shapes and gate signals and performs frequency detection. To change the input channels, you need to open the module and change the object [adc- 1 2]. Change the numbers 1 & 2 to the channel numbers you want to use.

CONTROLS:

INPUT - Boost the input signal from 0 - *4

AMP-SMOOTH - smooth the amplitude shape

GATE-THRESH - select threshold needed to extract gate signal

AMPLITUDE-/+ - attenuueverter to shape the AMPLITUDE output

INPUTS:

/none

OUTPUTS:

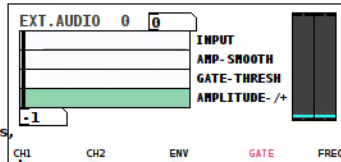
CH1 - left audio out

CH2 - right audio out

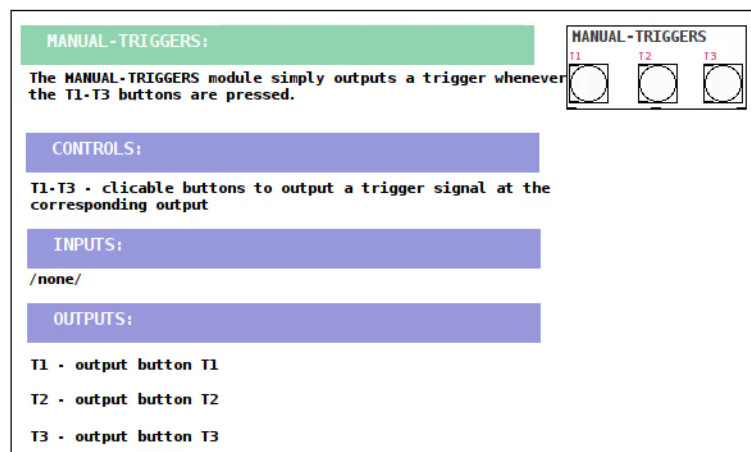
ENV - envelope output

GATE - gate output

FREQ - frequency detection output



manual_triggers



midi_in

MIDI-IN:

The "MIDI-IN" module receives MIDI note, gate and velocity signals from MIDI channel 1 and outputs them as Automatonism signals. Remember that all external MIDI gear must be connected before launching PD and selected in the PD MIDI Settings. If you want to read from another MIDI channel than channel 1, open the module and change the object [notein 1] to [notein yournumber].

CONTROLS:

/none/

INPUTS:

/none/

OUTPUTS:

NOTE - MIDI note output

GATE - MIDI gate output

VEL - MIDI velocity output

MIDI-IN

0

NOTE

GATE

VEL

midi_out

MIDI-OUT:

The MIDI-OUT module has been updated and improved for Automatonism v2.0 by Laurent Houdard.

The "MIDI-OUT" module takes a note input and a gate input to send Automatonism signals to external MIDI gear. Remember that all external MIDI gear must be connected before launching PD and selected in the PD MIDI Settings.

CONTROLS:

PITCH-OFFSET - offset the midi notes from the NOTE input

CHANNEL - select MIDI CHANNEL 1-16

INPUTS:

NOTE - midi pitch information input

GATE - midi gate information

OUTPUTS:

/none

NOTE

0

MIDI-OUT 0

PITCH-OFFSET

-64

CHANNEL

1

GATE

organelle_knobs

ORGANELLE-KNOBS:

The "organelle-knobs" module is a module that allows you to connect any of the four parameter knobs on the Critter & Guitari Organelle to cv inputs in the system. Rememebr that sliders on modules becomes offsets when cv is applied so make sure the parameter slider is set to zero to access the entire range.

CONTROLS:

none/

INPUTS:

none/

OUTPUTS:

Knob1 - output for Organelle knob1

Knob2 - output for Organelle knob2

Knob3 - output for Organelle knob3

Knob4 - output for Organelle knob4

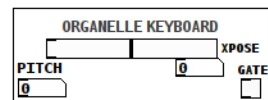
ORGANELLE KNOBS

Knob1	Knob2	Knob3
0	0	0

organelle_keyboard

ORGANELLE-KEYBOARD:

The "organelle-keyboard" module allows you to use the Critter & Guitari Organelle keyboard to cv inputs in the system. Remenebr that sliders on modules becomes offsets when cv is applied so make sure the parameter slider is set to zero to access the entire range.



CONTROLS:

XPOSE - transposes the octave of the Organelle keys

INPUTS:

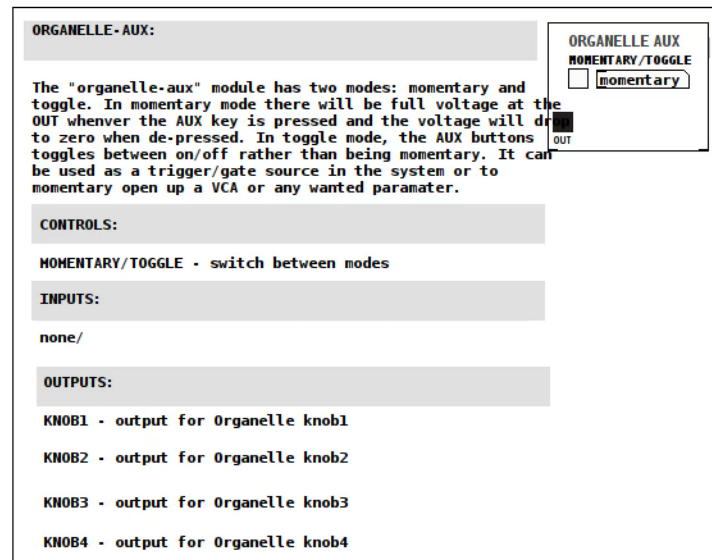
none/

OUTPUTS:

PITCH - outputs the pitch from the keyboard

GATE - outputs a trigger/gate signal when a key is pressed on the Organelle.

organelle_aux



organelle_output

ORGANELLE-OUTPUT:

The "ORGANELLE-OUTPUT" is used instead of the MAESTRO4 if you are exporting your patch to the Critter % Guitari Organelle.

CONTROLS:

CH1, CH2, CH3 & CH4 - controls gain of equivalent channel

PAN - control panning of equivalent channel

MASTER - controls the volume of the sum of channels 1-4

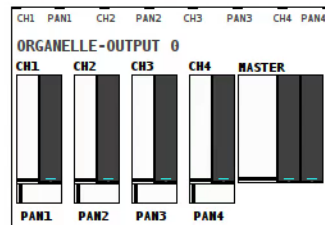
INPUTS:

CH1-4 - channel inputs

PAN1-4 - cv input for panning

OUTPUTS:

none/



Automatonism v2.1

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