



NONLINEARLABS

C15



Parameter Reference



NONLINEARLABS

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Comment

This document contains a parameter reference that is not up to date with the parameter extensions of the 2020 Update. Still, we think it will be a useful source of information until the latest version will be integrated into the C15 documentation.

Here a list of new parameters and other changes which are not included in the reference part:

Monophonic Modes

The voice allocation now supports monophonic playing with Lowest, Highest and Last Key priority, four different Legato modes and an adjustable Glide time. (Up to 24 unison voices can be layered.)

Six Macro Controls

The four existing MacroControls (**A**, **B**, **C**, **D**) are extended by **E** and **F**. All six Macros are also available in the MCView, a 2-D control surface for touch screens.

More Modulation Targets

Additional parameters (e.g. the Oscillator Phases) can be assigned to the MacroControls.

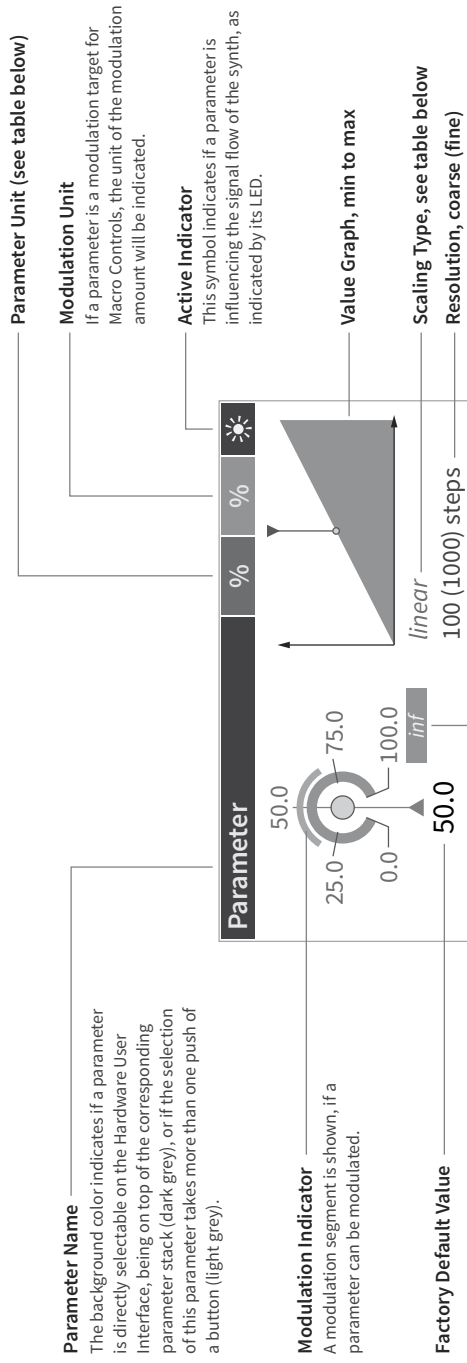
New Parameters

- Envelope A/B: Velocity amounts for the Decay 1 and Decay 2 times
- Envelope A/B: Elevate can raise the Breakpoint and Sustain levels for the timbre or the amplitude of an oscillator.
- Envelope C: Retrigger Hardness can control the starting point of mono voices
- Oscillator A/B: Reset can be disabled for free-running mode
- Tremolo effect as an extension of the Flanger
- Dual mode: Volume and Tune for Part **I** and **II**
- Layer mode: Fade From (key), Fade Range
- Feedback Mixer: more inputs for cross routing in LayerMode
- Output Mixer: signal split to both Dual mode effect chains
- Individual smoothing times for the six MacroControls
- Setting: Adjustable tuning reference: 440 +/-40 Hz

Extended Parameter Ranges and Resolutions

- Unison with up to 24 voices (before: 12)
- Bipolar Level Velocity amounts (for velocity crossfades)

Legend to signs and symbols



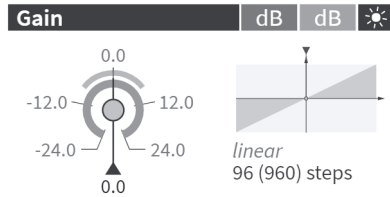
	Unit Reference
%	Amount-like parameters often use a normalized value range in percent (%), spanning from zero (or -100, when the parameter is bipolar) to 100 percent.
ms	Time-relevant parameters indicate their values in milliseconds (ms), meaning that 1000 ms equal one second.
st	Pitch-relevant parameters indicate their values in semitones (st). The interval of one semitone corresponds to two consecutive keys (at default key tracking). 12 semitones equal one octave.
ct	The Scale Offset parameters are also pitch-relevant, but show their values in cents (ct), offering more precision. 100 cents correspond to one semitone.
dB	Gain-relevant parameters show their values in decibel (dB), the common format of representing signal levels. Zero decibel correspond to 100 % signal, and a difference of 6 dB approaches a doubling of the signal.
dB_T	Attack and Release Velocity parameters come with a custom time-related format in decibel (dB_T). A full velocity will decrease the corresponding Envelope segment time by the indicated value in dB (6 dB approaching 50 % of the given time).
dB/st	Envelope Level Key Track parameters come with a custom gain-related format in decibel per semitone (dB/st). The corresponding gain will change according to the played key by the given amount of decibel per semitone.
Hz	The Flanger Rate parameter indicates its values in Hertz (Hz), the number of oscillations per second.
deg	Phase-related parameters indicate their values in degrees (deg), meaning that 360 degrees correspond to a full circle.

	Scaling Types
linear	Most parameters have a continuous, linear scaling type. This type is suitable when the displayed value is directly proportional to the control position and precision remains constant for the whole value range.
exponential	Envelope times and the Comb Filter Decay time have an exponential scaling type, covering a wide value range with decreasing precision. The displayed value doesn't rise by a constant value, but rather by a factor (20 per quarter).
parabolic	Some parameters (effect times for example) have a parabolic scaling type instead of being linear. The displayed value is proportional to the square of the control position in this case. The resulting behaviour of the parameter feels more realistic as if it would use linear scaling.
parabolic (gain)	The Mixer and Main Output levels come with a parabolic gain scaling type and show their value in decibels (dB). At minimum, they will mute the signal, at the center position, they will not affect the signal at all and at maximum, they will amplify the signal by 12 dB (~400%).
integer	Non-continuous parameters with discrete values (such as Union Voices and Scale Base Key) come with the integer scaling type. This is suitable for sufficiently low value ranges and results in a switch-like behaviour. The fine mode is not effective in this case.

Envelope A

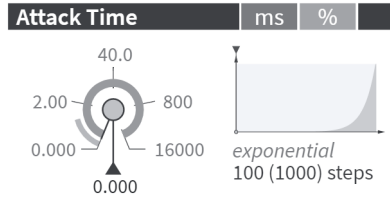
Attack Time	ms %	Time of the (polynomial) Attack segment in milliseconds.
Attack Velocity	dB_T	Velocity influence on the Attack time. The value represents the logarithmic amount of the reduction of the Attack time by high Note-On velocities.
Attack Curve	%	Curvature of the (polynomial) Attack segment. (negative: decelerating slope, zero: linear, positive: accelerating slope)
Decay 1 Time	ms %	Time of the first (linear) Decay segment in milliseconds.
Breakpoint Level	% %	Level of the Breakpoint between the two Decay segments.
Decay 2 Time	ms %	Time of the second (exponential) Decay segment in milliseconds.

Sustain Level	% % ☀	Sustain level (target of the second Decay segment).
Release Time	ms %	Time of the (exponential) Release segment in milliseconds (infinite at maximum).
Release Velocity	dB_T	Velocity influence on the Release time. The value represents the logarithmic amount of the reduction of the Release time by high Note-Off velocities.
Level Velocity	dB	Influence of the key velocity on the peak, breakpoint and sustain levels of the envelope [maximum dynamic range in dB].
Level Key Trk	dB/st	Key tracking of the envelope's peak, breakpoint and sustain levels [dB per semitone]. Positive values: higher levels for higher notes (+1.0 = +12 dB per octave). Negative values: lower levels for higher notes (-1.0 = -12 dB per octave). Origin at C3 = 60 semitones.
Time Key Trk	%	Key tracking of the attack, decay and release times. The value determines how much shorter the times get for higher notes.

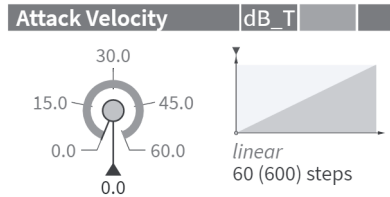


Gain [in dB] of the envelope signal. As this signal modulates the Oscillator and Shaper A, the Gain influences the level and the amount of phase modulation and distortion.

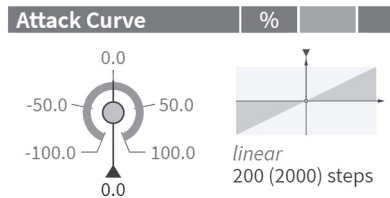
Envelope B



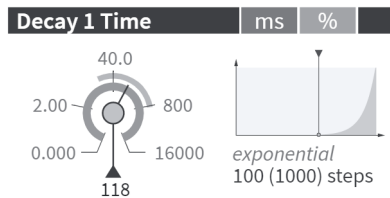
Time of the (polynomial) Attack segment in milliseconds.



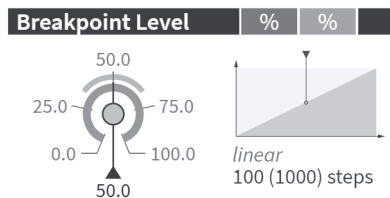
Velocity influence on the Attack time. The value represents the logarithmic amount of the reduction of the Attack time by high Note-On velocities.



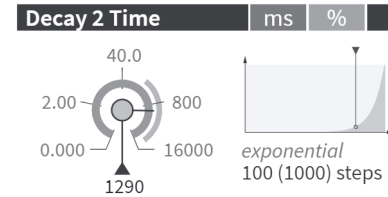
Curvature of the (polynomial) Attack segment. (negative: decelerating slope, zero: linear, positive: accelerating slope)



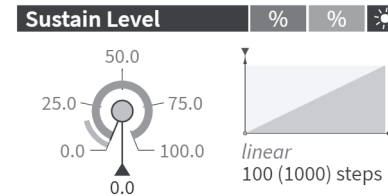
Time of the first (linear) Decay segment in milliseconds.



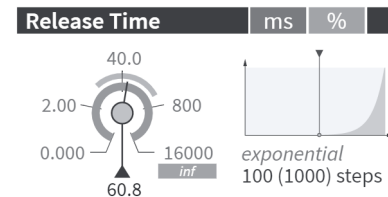
Level of the Breakpoint between the two Decay segments.



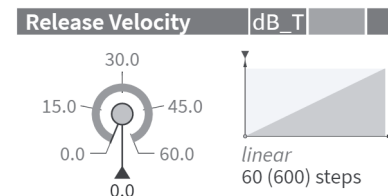
Time of the second (exponential) Decay segment in milliseconds.



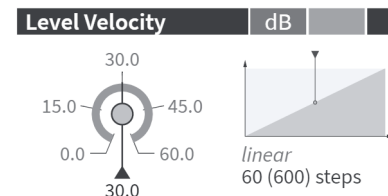
Sustain level (target of the second Decay segment).



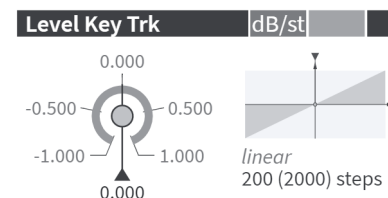
Time of the (exponential) Release segment in milliseconds (infinite at maximum).



Velocity influence on the Release time. The value represents the logarithmic amount of the reduction of the Release time by high Note-Off velocities.



Influence of the key velocity on the peak, breakpoint and sustain levels of the envelope [maximum dynamic range in dB].



Key tracking of the envelope's peak, breakpoint and sustain levels [dB per semitone].

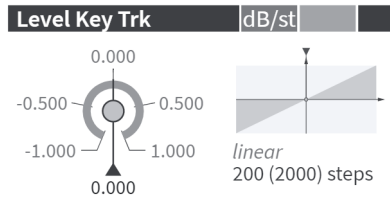
Positive values: higher levels for higher notes (+1.0 = +12 dB per octave). Negative values: lower levels for higher notes (-1.0 = -12 dB per octave). Origin at C3 = 60 semitones.

Time Key Trk	%		
			Key tracking of the attack, decay and release times. The value determines how much shorter the times get for higher notes.
Gain	dB	dB	
			Gain [in dB] of the envelope signal. As this signal modulates the Oscillator and Shaper B, the Gain influences the level and the amount of phase modulation and distortion.

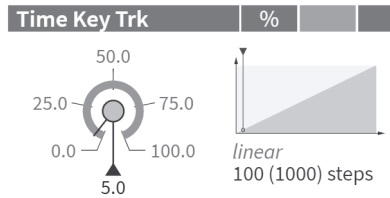
Envelope C

Attack Time	ms	%	
			Time of the (polynomial) Attack segment in milliseconds.
Attack Velocity	dB_T		
			Velocity influence on the Attack time. The value represents the logarithmic amount of the reduction of the Attack time by high Note-On velocities.
Attack Curve	%		
			Curvature of the (polynomial) Attack segment. (negative: decelerating slope, zero: linear, positive: accelerating slope)
Decay 1 Time	ms	%	
			Time of the first (linear) Decay segment in milliseconds.

Breakpoint Level	%	%	
			Level of the Breakpoint between the two Decay segments.
Decay 2 Time	ms	%	
			Time of the second (exponential) Decay segment in milliseconds.
Sustain Level	%	%	
			Sustain level (target of the second Decay segment).
Release Time	ms	%	
			Time of the (exponential) Release segment in milliseconds (infinite at maximum).
Release Velocity	dB_T		
			Velocity influence on the Release time. The value represents the logarithmic amount of the reduction of the Release time by high Note-Off velocities.
Level Velocity	dB		
			Influence of the key velocity on the peak, breakpoint and sustain levels of the envelope [maximum dynamic range in dB].

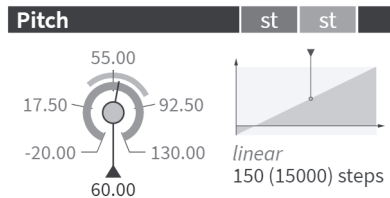


Key tracking of the envelope's peak, breakpoint and sustain levels [dB per semitone].
Positive values: higher levels for higher notes (+1.0 = +12 dB per octave).
Negative values: lower levels for higher notes (-1.0 = -12 dB per octave).
Origin at C3 = 60 semitones.

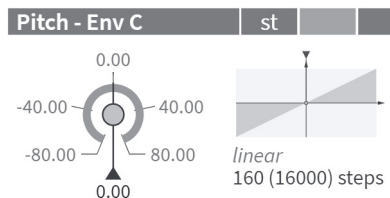


Key tracking of the attack, decay and release times. The value determines how much shorter the times get for higher notes.

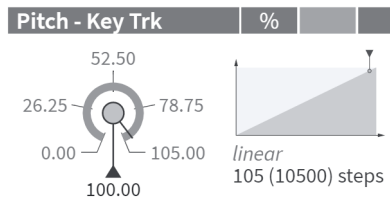
Oscillator A



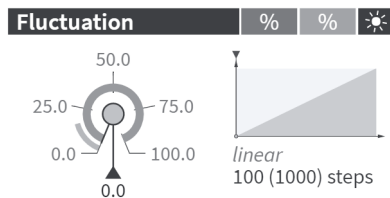
Pitch (logarithmic frequency) of Oscillator A at C3 (MIDI note 60) [in semitones, based on MIDI note numbers].
The range below zero is shaped in the way that -20 corresponds to 0 Hz.



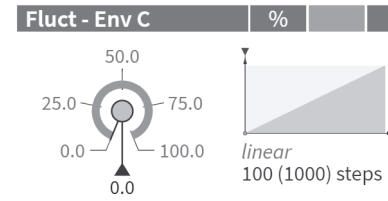
Amount of pitch modulation by Envelope C [in semitones].



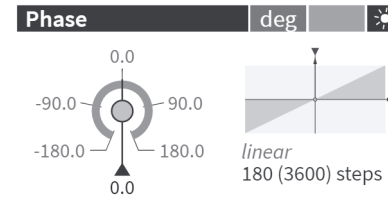
Key tracking of the oscillator pitch. It's the scaling factor between the key position (relative to C3 = 60 semitones) and the pitch of the oscillator. At 100.0 %, the pitch follows the equal-tempered scale. At values slightly larger than 100.0 %, the tuning will be stretched. At 50.0 %, a quartertone scale emerges and at 0.0 %, the pitch is constant for all keys.



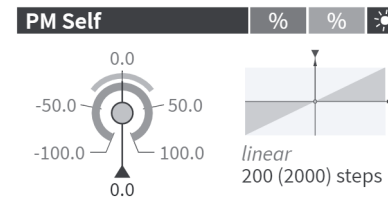
Fluctuation of the oscillator frequency.
If the parameter is set larger than 0.0 %, the frequency is changed at the beginning of each oscillation period by a random amount. At 100.0 %, the maximum frequency variation is +/- 95 %.



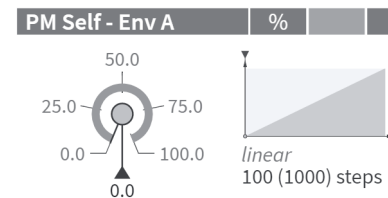
Amount of modulation of the frequency fluctuation by Envelope C. At zero, the modulation stays constant, at higher values, the envelope applies a time-variant attenuation.



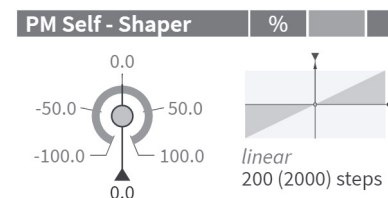
Start phase [in degrees]. The Oscillator will be set to this phase position with each Note-On.



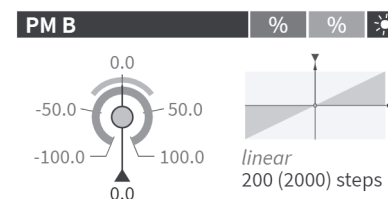
Amount of phase modulation by Oscillator & Shaper A (local feedback).



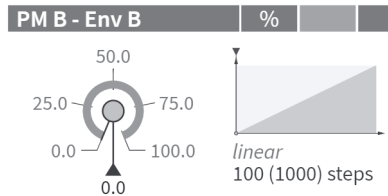
Envelope (A) amount for the phase modulation by Oscillator & Shaper A (local feedback).
At zero, the modulation stays constant, at higher values, the envelope applies a time-variant attenuation.



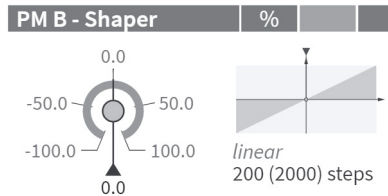
Mix amount of Shaper A in the signal being used for phase modulation (self). At zero, the output signal of Oscillator A is used. At negative values, the signal from the Shaper is inverted.



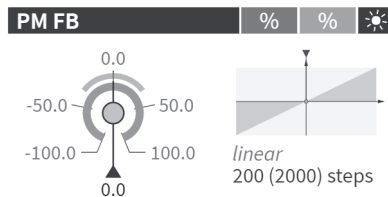
Amount of phase modulation by Oscillator & Shaper B (cross feedback).



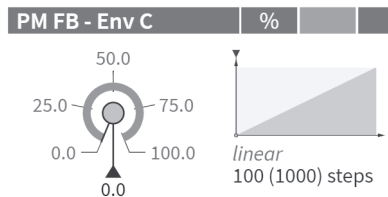
Envelope (B) amount for the phase modulation by Oscillator & Shaper B (cross feedback).
At zero, the modulation stays constant, at higher values, the envelope applies a time-variant attenuation.



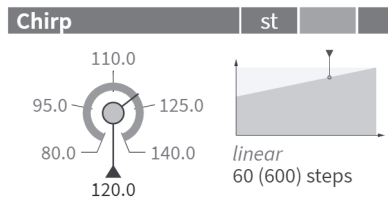
Mix amount of Shaper B in the signal being used for phase modulation (B -> A). At zero, the output signal of Oscillator B is used. At negative values, the signal from the Shaper is inverted.



Amount of phase modulation by the Feedback signal.

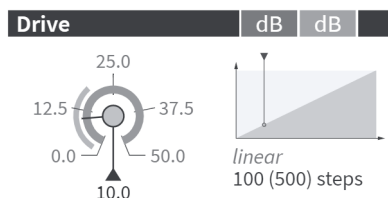


Envelope (C) amount for the phase modulation by the Feedback signal.
At zero, the modulation stays constant, at higher values, the envelope applies a time-variant attenuation.

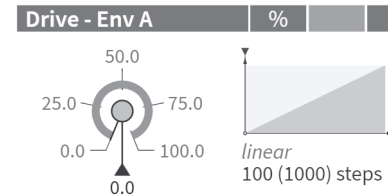


Cutoff [in semitones] of the lowpass in the phase modulation signal path that can be applied to reduce the level and frequency of "chirping" appearing at higher amounts of self modulation, cross-feedback through the other oscillator, or global feedback.

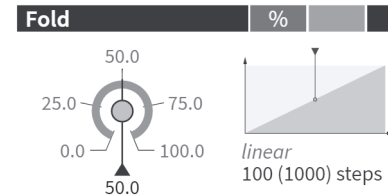
Shaper A



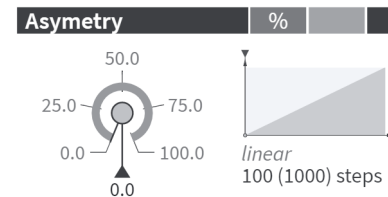
Input gain [in dB] of the sine shaper stage. Higher gains will create more distortion and harmonics.



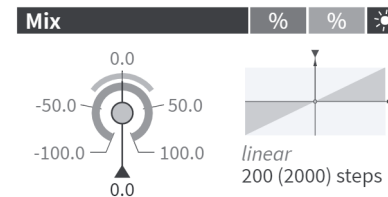
Envelope (A) amount for the Drive factor.
At zero, the gain stays constant, at higher values, the envelope applies a time-variant attenuation.



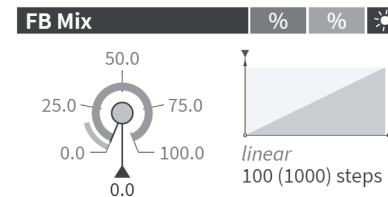
Amount of folding back of the shaper curve for high input amplitudes.
0.0 %: flat saturation, no folding
100 %: fully folded back (periodic sine curve)
A higher amount of folding leads to a softer but more nasal sound.



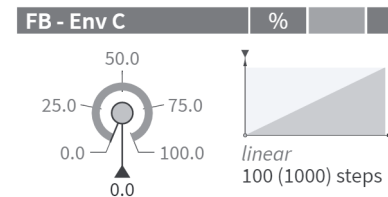
Asymmetry of the shaper curve, generating even (2nd, 4th, ...) harmonics. At higher values it becomes a parabolic curve that shifts the frequency of the fundamental to its double.



Mix amount of Shaper A in the signal sent to the Filters and to the Output Mixer. At zero, it is the input signal of the Shaper - behind FB Mix. At negative values, the signal from the Shaper is inverted.

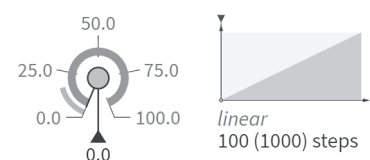


Crossfades between Oscillator & Shaper A (at zero) and the Feedback signal for the signal A.



Envelope (C) amount for the Feedback Mix. At zero, only a Gate signal is applied, at higher values, Envelope C is faded in.

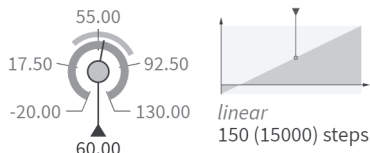
Ring Mod % % ☀



Mix amount of the ring modulation between both Oscillators & Shapers.

Oscillator B

Pitch st st



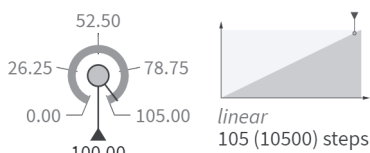
Pitch (logarithmic frequency) of Oscillator B at C3 (MIDI note 60) [in semitones, based on MIDI note numbers]. The range below zero is shaped in the way that -20 corresponds to 0 Hz.

Pitch - Env C st



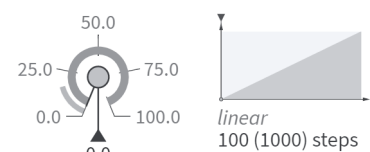
Amount of pitch modulation by Envelope C [in semitones].

Pitch - Key Trk %



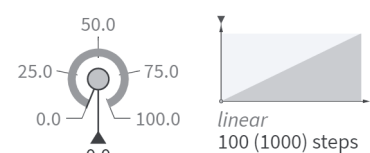
Key tracking of the oscillator pitch. It's the scaling factor between the key position (relative to C3 = 60 semitones) and the pitch of the oscillator. At 100.0 %, the pitch follows the equal-tempered scale. At values slightly larger than 100.0 %, the tuning will be stretched. At 50.0 %, a quartertone scale emerges and at 0.0 %, the pitch is constant for all keys.

Fluctuation % % ☀



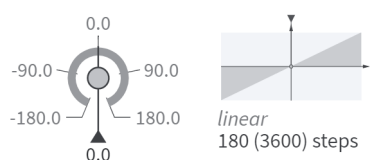
Fluctuation of the oscillator frequency. If the parameter is set larger than 0.0 %, the frequency is changed at the beginning of each oscillation period by a random amount. At 100.0 %, the maximum frequency variation is +/- 95 %.

Fluct - Env C %




Amount of modulation of the frequency fluctuation by Envelope C. At zero, the modulation stays constant, at higher values, the envelope applies a time-variant attenuation.

Phase deg ☀



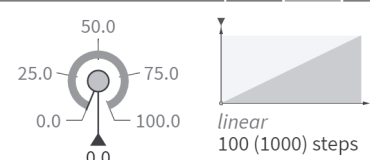
Start phase [in degrees]. The Oscillator will be set to this phase position with each Note-On.

PM Self % % ☀




Amount of phase modulation by Oscillator & Shaper B (local feedback).

PM Self - Env B %




Envelope (B) amount for the phase modulation by Oscillator & Shaper B (local feedback). At zero, the modulation stays constant, at higher values, the envelope applies a time-variant attenuation.

PM Self - Shaper %



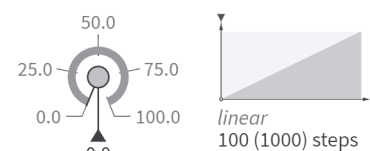
Mix amount of Shaper B in the signal being used for phase modulation (self). At zero, the output signal of Oscillator B is used. At negative values, the signal from the Shaper is inverted.

PM A % % ☀

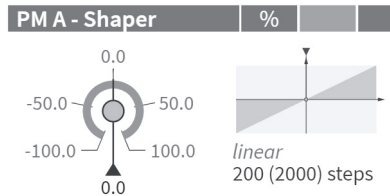


Amount of phase modulation by Oscillator & Shaper A (cross feedback).

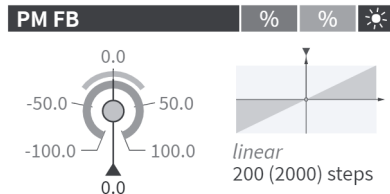
PM A - Env A %



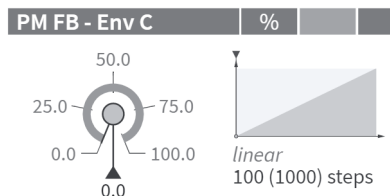
Envelope (A) amount for the phase modulation by Oscillator & Shaper A (cross feedback). At zero, the modulation stays constant, at higher values, the envelope applies a time-variant attenuation.



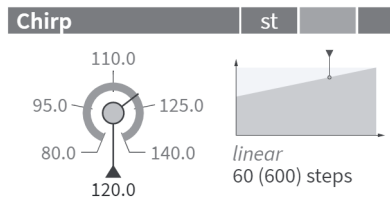
Mix amount of Shaper A in the signal being used for phase modulation (A → B). At zero, the output signal of Oscillator A is used. At negative values, the signal from the Shaper is inverted.



Amount of phase modulation by the Feedback signal.

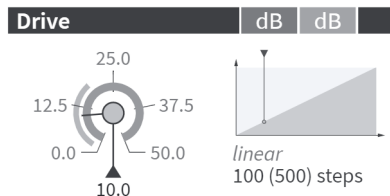


Envelope (C) amount for the phase modulation by the Feedback signal.
At zero, the modulation stays constant, at higher values, the envelope applies a time-variant attenuation.

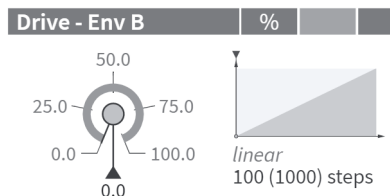


Cutoff [in semitones] of the lowpass in the phase modulation signal path that can be applied to reduce the level and frequency of “chirping” appearing at higher amounts of self modulation, cross-feedback through the other oscillator, or global feedback.

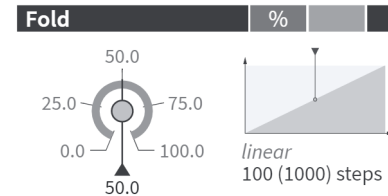
Shaper B



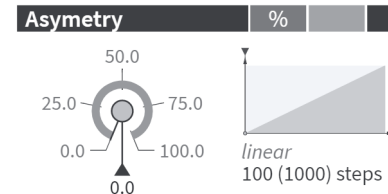
Input gain [in dB] of the sine shaper stage. Higher gains will create more distortion and harmonics.



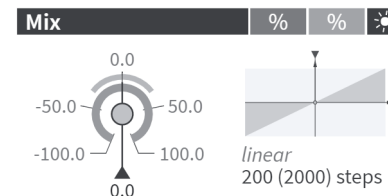
Envelope (B) amount for the Drive factor.
At zero, the gain stays constant, at higher values, the envelope applies a time-variant attenuation.



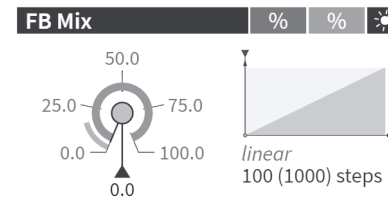
Amount of folding back of the shaper curve for high input amplitudes.
0.0 %: flat saturation, no folding
100 %: fully folded back (periodic sine curve)
A higher amount of folding leads to a softer but more nasal sound.



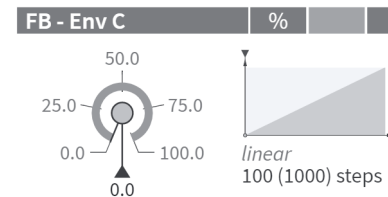
Asymmetry of the shaper curve, generating even (2nd, 4th, ...) harmonics. At higher values it becomes a parabolic curve that shifts the frequency of the fundamental to its double.



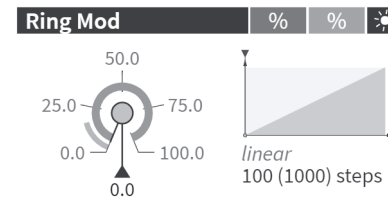
Mix amount of Shaper B in the signal sent to the Filters and to the Output Mixer. At zero, it is the input signal of the Shaper - behind FB Mix. At negative values, the signal from the Shaper is inverted.



Crossfades between Oscillator & Shaper B (at zero) and the Feedback signal for the signal B.



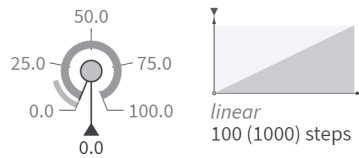
Envelope (C) amount for the Feedback Mix. At zero, only a Gate signal is applied, at higher values, Envelope C is faded in.



Mix amount of the ring modulation between both Oscillators & Shapers.

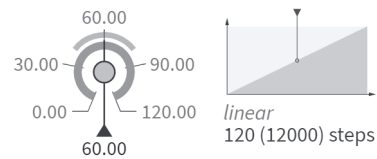
Comb Filter

A - B



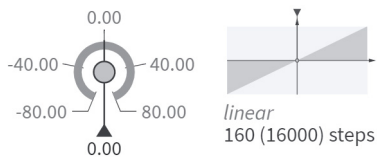
The signal for the Comb Filter as a crossfade between the outputs of Oscillator & Shaper A and Oscillator & Shaper B.

Pitch



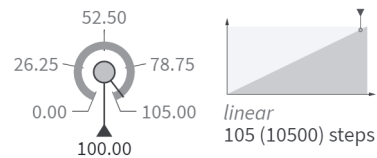
Coarse pitch of the Comb Filter (delay) at C3 (MIDI note 60) [in semitones, based on MIDI note numbers].

Pitch - Env C



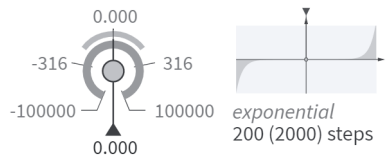
Amount of modulation of the Comb Filter pitch by Envelope C [in semitones].

Pitch - Key Trk



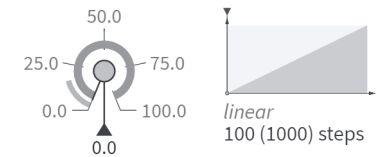
Key scaling of the tuning of the Comb Filter (delay).
0.0 %: same tuning for all keys
100.0 %: full tracking with keys, origin at C3 = 60 semitones

Decay



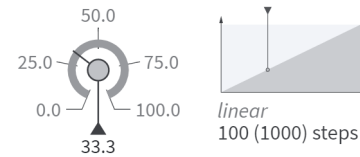
Amount of the internal feedback noticeable as the decay time of the impulse response [logarithmic scaling]. At negative values, the feedback signal is inverted, which shifts the fundamental resonance down by one octave.

Decay - Gate



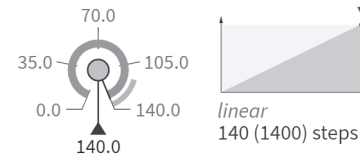
Amount of gating applied to the decay time. It reduces the decay time when the key is released (Note-off).
0.0 % same decay time in the release phase
100.0 %: the decay time is set to zero at the release of the key

Decay - Key Trk



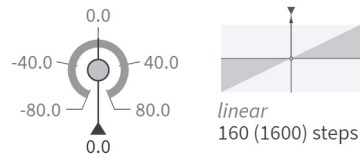
Key scaling of the decay time.
0.0 %: equal time for all keys
100.0 %: shortening to half time per octave, origin at C3 = 60 semitones

Allpass - Tune



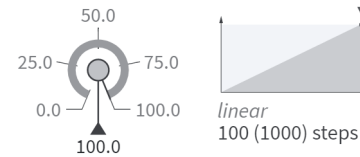
Center frequency of the 2-pole allpass filter [in semitones]. It is the frequency where the phase is shifted by 180 degrees. The allpass is in series with the delay. At the maximum position (140 semitones), the allpass has no effect.

Allpass - Env C



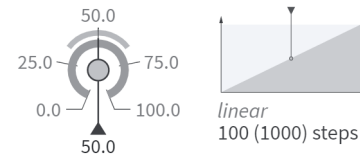
Amount of modulation of the allpass center frequency by Envelope C [in semitones].

Allpass - Key Trk



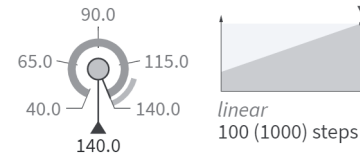
Key scaling of the allpass filter frequency.
0.0 %: same tuning for all keys
100.0 %: full tracking with the keys, origin at C3 = 60 semitones

Allpass - Reson



Resonance of the 2-pole allpass. Determines how much the phase shift increases around the center frequency.

Hi Cut



Cutoff frequency [in semitones] of the lowpass filter that damps the higher frequencies in the comb filter's signal path.

Hi Cut - Env C st st

Amount of modulation of the cutoff frequency of the lowpass filter by Envelope C [in semitones].

Hi Cut - Key Trk %

Key scaling of the lowpass cutoff frequency.
0.0 %: same cutoff for all keys
100.0 %: full tracking with the keys, origin at C3 = 60 semitones

PM % %

Amount of (phase) modulation of the comb filter pitch (delay time) by the output signals of Oscillator & Shaper A and B, with the PM A-B knob determining their mix.

PM - A-B %

The (phase) modulation signal for the Comb Filter as a crossfade between the outputs of Oscillator & Shaper A and Oscillator & Shaper B.

State Variable Filter

A - B % %

The signal for the State Variable Filter as a crossfade between the outputs of Oscillator & Shaper A and Oscillator & Shaper B.

Comb Mix % %

The input signal for the State Variable Filter as mix of the signals from the A-B mixer and from the Comb Filter. Negative mix amounts will create different Comb Filter (cancellation) effects.

Cutoff st st

Static value of the filter cutoff frequency at C3 [in semitones], applies to both stages of the filter. The offsets between their individual cutoffs is controlled by "Spread".

Cutoff - Env C st st

Amount of cutoff modulation by Envelope C [in semitones].

Cutoff - Key Trk %

Key scaling of the filter cutoffs.
0.0 %: no influence
100.0 %: full tracking with the keys, origin at C3 = 60 semitones

Resonance % %

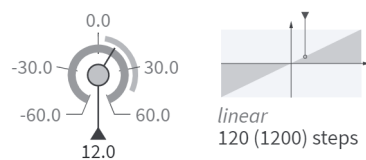
Amount of filter resonance, creating peaks at the cutoff frequencies.

Reson - Env C %

Amount of resonance modulation by Envelope C.

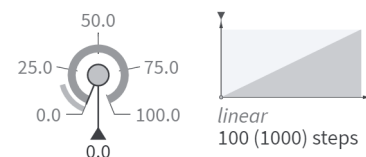
Reson - Key Trk st st

Key scaling of the filter resonance.
0.0 %: no influence
100.0 %: full tracking with the keys, origin at C3 = 60 semitones

Spread

Amount of splitting of the cutoffs of the two 2-pole filters. Half of the value is applied as a positive offset to the adjusted cutoff for the first stage and as a negative offset for the second stage [in semitones].

The split reduces the strong resonance peak in the 4-pole mode and allows filter curves with two formants. In band-pass/bandreject mode it controls the width of the band.

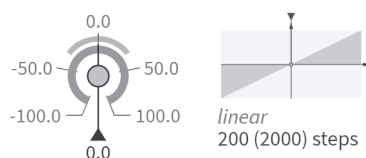
L - B - H

Crossfades between the lowpass, bandpass and highpass outputs of the two filter stages.

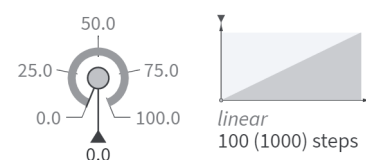
(first stage -> second stage: LP -> LP, BP -> LP, HP -> LP, HP -> BP, HP -> HP)

Parallel

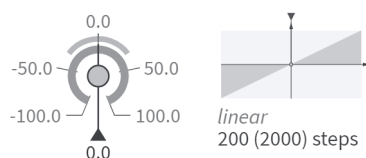
At zero, the two 2-pole filter stages are in series forming a 4-pole filter. Positive values crossfade to a parallel structure where the filter outputs are added. In bandpass mode (L-B-H = 50.0%), the parallel structure with a negative Spread works as a band-reject filter. Negative values also crossfade to a parallel structure, but here the lower filter is subtracted, which leads to phase cancellations.

FM

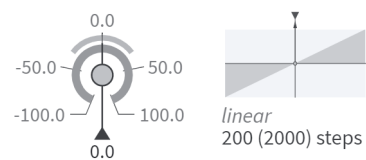
Amount of modulation of the cutoff frequencies by the output signals of Oscillator & Shaper A and B, with the FM A-B knob determining their mix. The amount is relative to the cutoff frequency.

FM - A-B

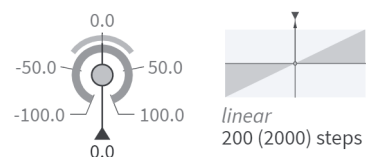
The signal for the FM (cutoff frequency modulation) of the State Variable Filter as a crossfade between the outputs of Oscillator & Shaper A and Oscillator & Shaper B.

Feedback Mixer**Comb Filter**

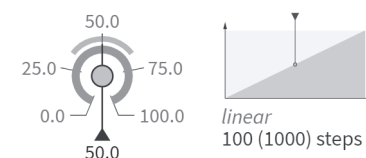
Feedback mix factor for the output of the Comb Filter.

SV Filter

Feedback mix factor for the output of the State Variable Filter.

Effects

Feedback mix factor for the output of the Effects chain. The reverb amount in the feedback can be set by the "Reverb Amount" fader independantly. Since the signal is monophonic, such feedback will cause intermodulation between the voices.

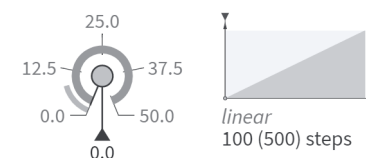
Reverb Amount

Controls the amount of reverb in the feedback independantly from the Mix in the Reverb section.

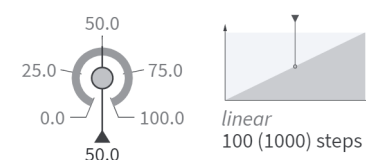
0.0 %: dry, no reverb signal

50.0 %: mix of 50 % dry and 50 % wet signal

100.0 %: wet, 100 % reverb signal

Drive

Input gain [in dB] of the sine shaper stage. Higher gains will create more distortion and harmonics.

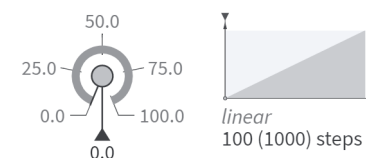
Fold

Amount of folding back of the shaper curve for high input amplitudes.

0.0 %: flat saturation, no folding

100.0 %: fully folded back (periodic sine curve)

A higher amount of folding leads to a softer but more nasal sound.

Asymetry

Asymmetry of the shaper curve, generating even (2nd, 4th, ...) harmonics. At higher values, it becomes a parabolic curve that shifts the frequency of the fundamental to its double.

Level	dB	%	
			Level of the (global) Feedback mix [in dB].
Level - Key Trk	dB/st		
			Key scaling of the feedback level [in dB per semitone]. positive values: higher level for higher notes (+ 1.0 = + 12 dB per octave) negative values: lower level for higher notes (- 1.0 = - 12 dB per octave) origin at C3 = 60 semitones

Output Mixer

A - Level	%	%	
			Output mix factor for the signal from Oscillator & Shaper A.
A - Pan	%		
			Output mix factor for the signal from Oscillator & Shaper A.
B - Level	%	%	
			Output mix factor for the signal from Oscillator & Shaper B.
B - Pan	%		
			Output mix factor for the signal from Oscillator & Shaper B.

Comb - Level	%	%	
			Output mix factor for the signal from the Comb Filter.
Comb - Pan	%		
			Pan position of the signal from the Comb Filter in the output mix.
SV Filter - Level	%	%	
			Output mix factor for the signal from the State Variable Filter.
SV Filter - Pan	%		
			Pan position of the signal from the State Variable Filter in the output mix.
Drive	dB	dB	
			Input gain [in dB] of the sine shaper stage. Higher gains will create more distortion and harmonics.
Fold	%		
			Amount of folding back of the shaper curve for high input amplitudes. 0.0 %: flat saturation, no folding 100.0 %: fully folded back (periodic sine curve) A higher amount of folding leads to a softer but more nasal sound.

Asymetry %

50.0
25.0 75.0
0.0 100.0
0.0
linear
100 (1000) steps

Asymetry of the shaper curve, generating even (2nd, 4th, ...) harmonics. At higher values, it becomes a parabolic curve that shifts the frequency of the fundamental to its double.

Level dB %

0.000
-12.0 7.04
-inf 12.0
-4.77
(parabolic) gain
100 (1000) steps

Master output level [in dB] of the synth. The output signal will be processed by the (monophonic) chain of effects.

Key Pan %

50.0
25.0 75.0
0.0 100.0
0.0
linear
100 (1000) steps

Amount of key panning for each note (referring to the key position relative to the center at C3 = 60 semitones). At zero, each note is centered before being passed to the panning section in the mix. At maximum, low notes will be panned to the left, high notes to the right.

Flanger

Rate Hz %

2.50
0.625 5.63
0.000 10.0
1.00
parabolic
100 (1000) steps

Frequency of the LFO and rate of the envelope. Both can be modulation sources for the delay times and the allpass center frequencies.

Envelope %

50.0
25.0 75.0
0.0 100.0
0.0
linear
100 (1000) steps

Crossfades between the signals of the LFO and the envelope as modulation sources for the delay times and the allpass center frequencies.

Phase deg

90.0
45.0 135.0
0.0 180.0
90.0
linear
180 (1800) steps

Phase offset between the LFO signals modulating the delay times in the left and the right channel.

Time Mod % %

0.000
-25.0 25.0
-100 100
0.000
parabolic
200 (2000) steps

Relative amount of the modulation of the delay times by the LFO and/or the envelope.

Time ms %

12.5
3.14 28.2
0.000 50.0
5.02
parabolic
125 (1250) steps

Mean value of the delay times in the left and right channel. The "Stereo" parameter allows to create a time offset between the channels. When the time of the Flanger is set to zero, the overall effect is determined by the phase shifting of the allpass.

Stereo %

0.0
-25.0 25.0
-50.0 50.0
0.0
linear
100 (1000) steps

Sets the ratio between the delay times of the left and of the right channel [the value shows the offset to 100.0 %]. In the center position, the offset is zero and both delay times are equal.

Allpass Mod % %

0.0
-50.0 50.0
-100.0 100.0
0.0
linear
200 (2000) steps

Relative amount of the modulation of the allpass center frequencies by the LFO and/or the envelope. The allpass creates the effect of a phaser.

Allpass Tune st st

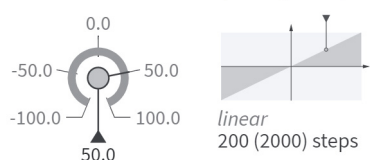
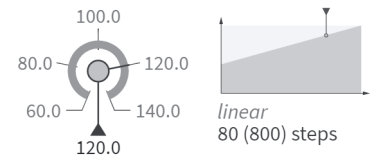

70.0
35.0 105.0
0.0 140.0
140.0
linear
140 (1400) steps

Mean center frequency of the 4-pole allpass filters which are in series with the delays. Their frequency-dependant phase shifts can create a "Phaser" effect. The phase shift is minimized by setting this control to its maximum. When the time of the flanger is set to zero, the allpass becomes the dominant part.

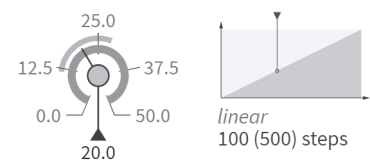
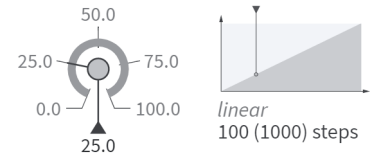
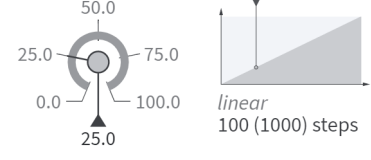
Feedback % %

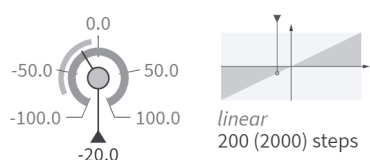
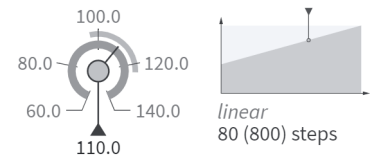
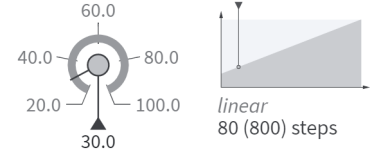
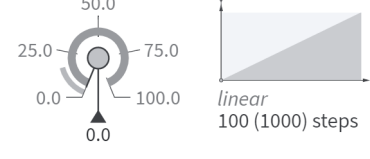
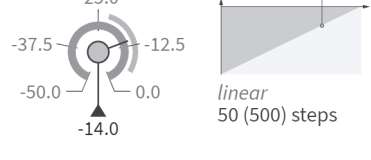
0.0
-50.0 50.0
-100.0 100.0
0.0
linear
200 (2000) steps

Amount of the internal feedback. At negative values, the feedback is inverted and will emphasize other frequencies than in the non-inverted mode.

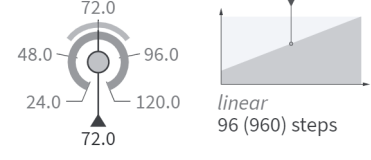
Cross FB	%			Amount of the cross feedback between the left and the right channel, increasing the complexity of the resulting signal. At negative values, the cross feedback is inverted and will emphasize other frequencies than in the non-inverted mode. 0.0 %: no cross feedback +/- 100.0 %: all feedback is cross feedback
				
Hi Cut	st			Cutoff frequency [in semitones] of the filter that damps the higher frequencies of the delayed signal.
				
Mix	%	%	☀	Crossfades between the dry signal and the delayed signal. At negative values, the delayed signal is inverted.
				

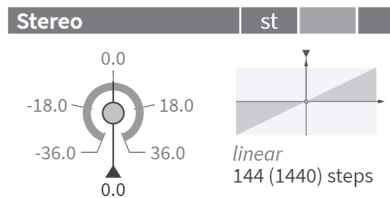
Cabinet

Drive	dB	dB		Gain [in dB] for the input signal. Higher gains will increase the amount of distortion/saturation.
				
Fold	%			Amount of folding back of the shaper curve for high input amplitudes. 0.0 %: flat saturation, no folding 100.0 %: fully folded back (periodic sine curve) A higher amount of folding leads to a softer but more nasal sound.
				
Asymetry	%			Asymmetry of the shaper curve, generating even (2nd, 4th, ...) harmonics. At higher values, it becomes a parabolic curve that shifts the frequency of the fundamental to its double.
				

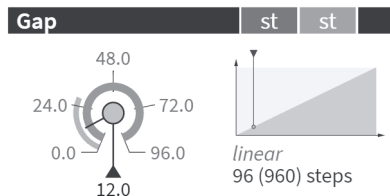
Tilt	%	%		Controls two inverted shelving EQs before and after the distortion stage. negative: more distortions at high frequencies positive: more distortions at low frequencies
				
Hi Cut	st	st		Cutoff frequency [in semitones] of the lowpass filter at the output.
				
Lo Cut	st			Cutoff frequency [in semitones] of the highpass filter at the input.
				
Mix	%	%	☀	Crossfades between the dry signal and the saturated signal.
				
Cab Level	dB	dB		Output level [in dB] of the saturation effect before it is mixed with the dry signal.
				

Gap Filter

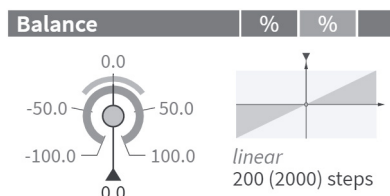
Center	st	st		Shifts the mean cutoff frequency of both 4-pole filters on both channels up or down [in semitones].
				



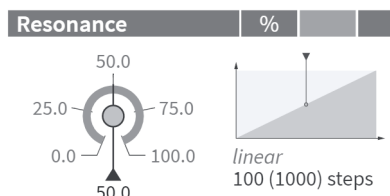
Sets the difference between the center frequencies of the left and of the right channel [in semitones].



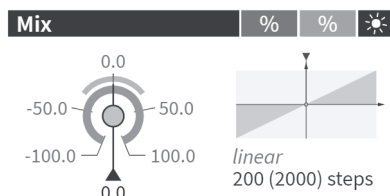
Offset between the cutoffs of the lowpass and the highpass [in semitones]. Since the two filters are running in parallel and their output signals are mixed, the result of a positive gap is a band rejection. With a negative gap, the pass bands are overlapping so that all frequencies can pass and the resonances emphasize the cutoff frequencies.



Balance between the levels of the ranges above and below the gap. In the center position both ranges are equally weighted. Negative values boost the lower range and attenuate the higher range, positive values have the opposite effect.

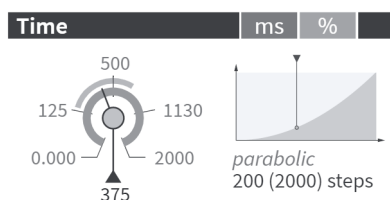


Resonance of both filters. Higher values create two resonance peaks at the upper and lower end of the gap.

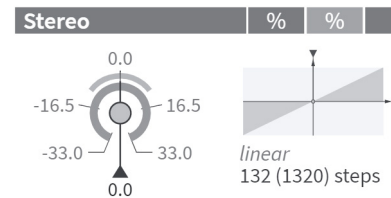


Crossfades between the dry signal and the filtered signal. At positive values, the filter runs in parallel (band reject) mode, at negative values, it is a bandpass filter (in serial mode).

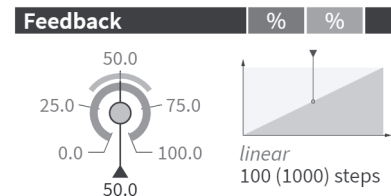
Echo



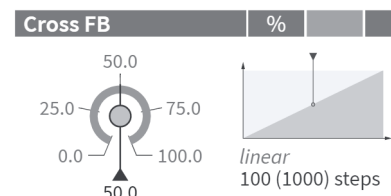
Mean delay time [in milliseconds]. (As there can be an offset between the left and right channel, this control shows the mean time.)



Sets the ratio between the delay times of the left and of the right channel [the value shows the offset to 100.0 %]. In the center position, the offset is zero and both delay times are equal.

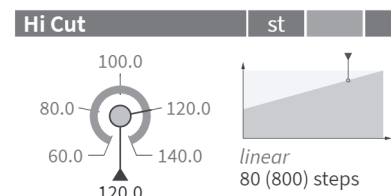


Amount of internal feedback.

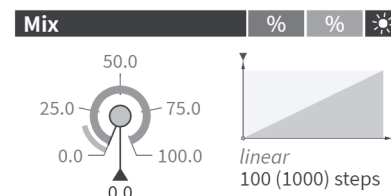


Amount of the cross feedback between the left and right channel.

0.0 %: no cross feedback
100.0 %: all feedback is cross feedback

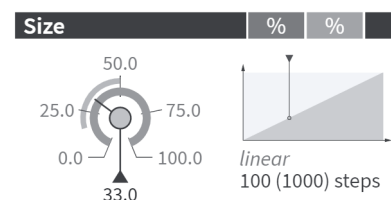


Cutoff frequency [in semitones] of the filter that damps the higher frequencies of the delayed signal.

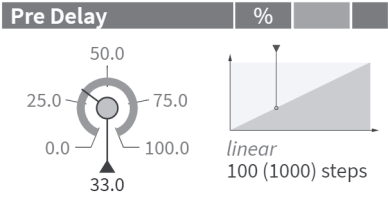
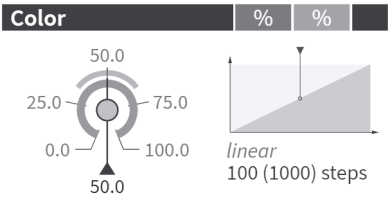
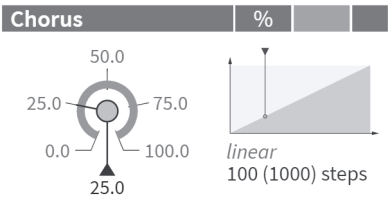
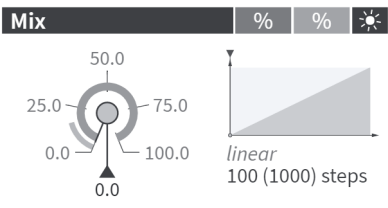


Crossfades between the dry signal and the delayed signal.

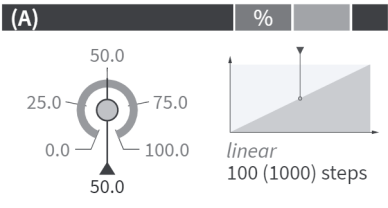
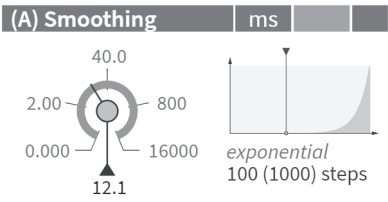
Reverb

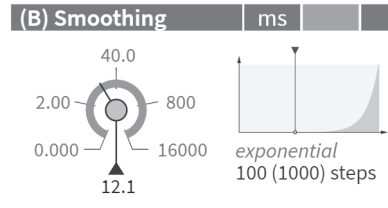
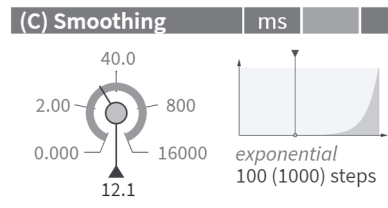
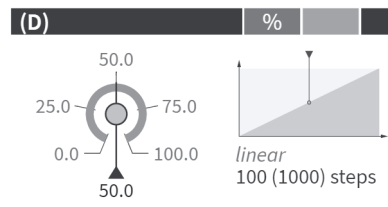
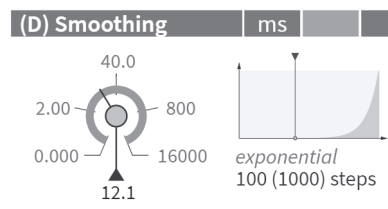


The room size and reverb time are set here.

<p>Pre Delay %</p> 	<p>Pre delay time, shifting the late reflections. This has a profound effect on the perceived room size.</p>
<p>Color % %</p> 	<p>Cutoff of the filter that damps the lower frequencies of the reverberation signal.</p>
<p>Chorus %</p> 	<p>This controls the internal modulation of the reverberation delays. At higher amounts, there is more movement and the diffusion is smoother. At lower amounts, the reverb becomes more static and metallic.</p>
<p>Mix % % ☀</p> 	<p>Crossfades between the dry signal and the reverberation signal.</p>

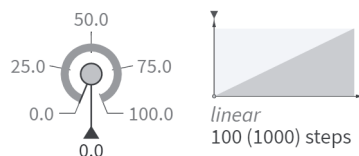
Macros

<p>(A) %</p> 	<p>The info text of Macro Control A is user-definable and defaults to an empty string.</p>
<p>(A) Smoothing ms</p> 	<p>Sets the smoothing time [in milliseconds] for modulations on target parameters assigned to Macro Control A. At zero, assigned target parameters will follow the Macro Control immediately.</p>

<p>(B) %</p> 	<p>The info text of Macro Control B is user-definable and defaults to an empty string.</p>
<p>(B) Smoothing ms</p> 	<p>Sets the smoothing time [in milliseconds] for modulations on target parameters assigned to Macro Control B. At zero, assigned target parameters will follow the Macro Control immediately.</p>
<p>(C) %</p> 	<p>The info text of Macro Control C is user-definable and defaults to an empty string.</p>
<p>(C) Smoothing ms</p> 	<p>Sets the smoothing time [in milliseconds] for modulations on target parameters assigned to Macro Control C. At zero, assigned target parameters will follow the Macro Control immediately.</p>
<p>(D) %</p> 	<p>The info text of Macro Control D is user-definable and defaults to an empty string.</p>
<p>(D) Smoothing ms</p> 	<p>Sets the smoothing time [in milliseconds] for modulations on target parameters assigned to Macro Control D. At zero, assigned target parameters will follow the Macro Control immediately.</p>

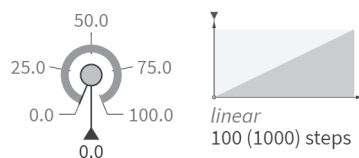
Hardware Sources

Pedal 1



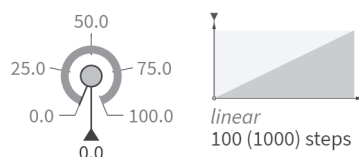
This control is the parameter representation of the “Pedal 1” Hardware Source. If a pedal is connected, it will directly follow the external control. If the source’s return behavior is set to “Center”, the parameter will be bipolar.

Pedal 2



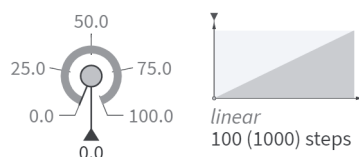
This control is the parameter representation of the “Pedal 2” Hardware Source. If a pedal is connected, it will directly follow the external control. If the source’s return behavior is set to “Center”, the parameter will be bipolar.

Pedal 3



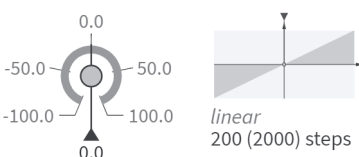
This control is the parameter representation of the “Pedal 3” Hardware Source. If a pedal is connected, it will directly follow the external control. If the source’s return behavior is set to “Center”, the parameter will be bipolar.

Pedal 4



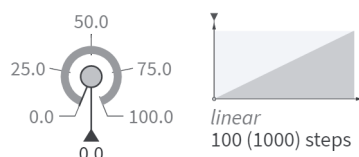
This control is the parameter representation of the “Pedal 4” Hardware Source. If a pedal is connected, it will directly follow the external control. If the source’s return behavior is set to “Center”, the parameter will be bipolar.

Bender



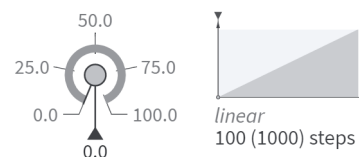
This control is the parameter representation of the “Bender” Hardware Source. It will directly follow the Bender.

Aftertouch



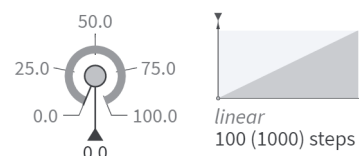
This control is the parameter representation of the “Aftertouch” Hardware Source. It will directly follow Aftertouch gestures.

Ribbon 1



This control is the parameter representation of the “Ribbon 1” Hardware Source. It will directly follow the Ribbon and vice versa. If the source’s return behavior is set to “Center”, the parameter will be bipolar.

Ribbon 2



This control is the parameter representation of the “Ribbon 2” Hardware Source. It will directly follow the Ribbon and vice versa. If the source’s return behavior is set to “Center”, the parameter will be bipolar.

Hardware Amounts

Ped 1 to (A)



Determines the influence of the “Pedal 1” Hardware Source on Macro Control A. If the Pedal is a returning Hardware Source, the amount can be adjusted continuously, otherwise it will be switch-like. Continuous amounts can be set [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).

Ped 1 to (B)



Determines the influence of the “Pedal 1” Hardware Source on Macro Control B. If the Pedal is a returning Hardware Source, the amount can be adjusted continuously, otherwise it will be switch-like. Continuous amounts can be set [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).

Ped 1 to (C)

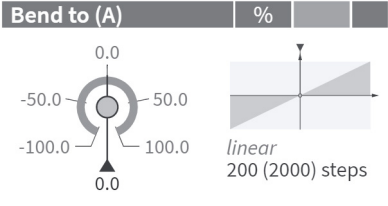
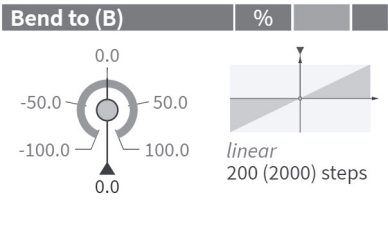
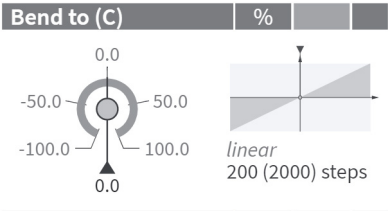
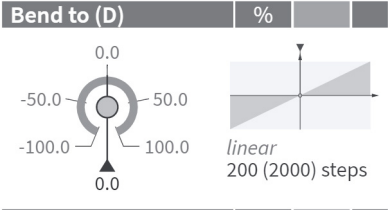
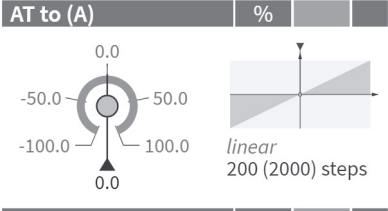
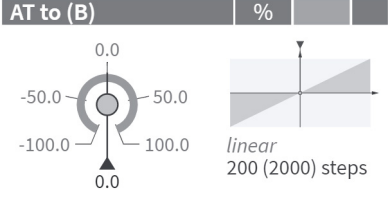


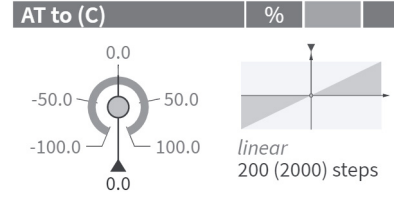
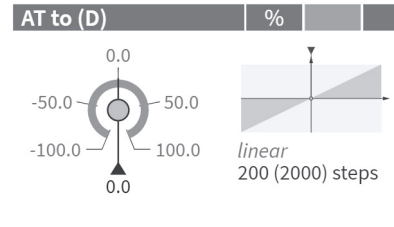
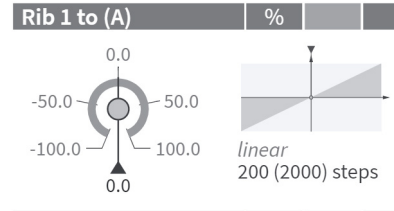
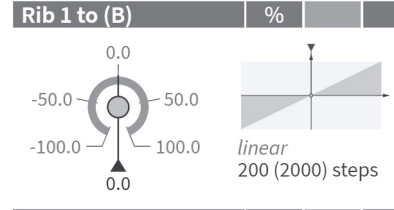
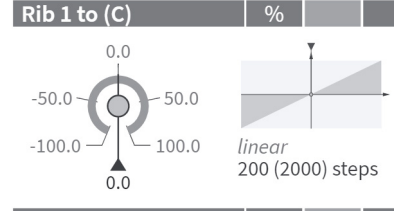
Determines the influence of the “Pedal 1” Hardware Source on Macro Control C. If the Pedal is a returning Hardware Source, the amount can be adjusted continuously, otherwise it will be switch-like. Continuous amounts can be set [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).

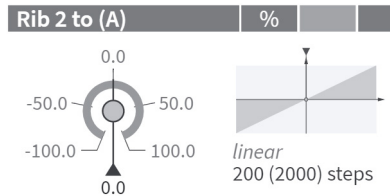
Ped 1 to (D)



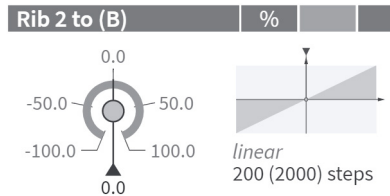
Determines the influence of the “Pedal 1” Hardware Source on Macro Control D. If the Pedal is a returning Hardware Source, the amount can be adjusted continuously, otherwise it will be switch-like. Continuous amounts can be set [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).

Bend to (A) % <input type="checkbox"/> 	<p>Determines the influence of the “Bender” Hardware Source on Macro Control A.</p> <p>The amount can be adjusted continuously [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).</p>
Bend to (B) % <input type="checkbox"/> 	<p>Determines the influence of the “Bender” Hardware Source on Macro Control B.</p> <p>The amount can be adjusted continuously [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).</p>
Bend to (C) % <input type="checkbox"/> 	<p>Determines the influence of the “Bender” Hardware Source on Macro Control C.</p> <p>The amount can be adjusted continuously [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).</p>
Bend to (D) % <input type="checkbox"/> 	<p>Determines the influence of the “Bender” Hardware Source on Macro Control D.</p> <p>The amount can be adjusted continuously [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).</p>
AT to (A) % <input type="checkbox"/> 	<p>Determines the influence of the “Aftertouch” Hardware Source on Macro Control A.</p> <p>The amount can be adjusted continuously [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).</p>
AT to (B) % <input type="checkbox"/> 	<p>Determines the influence of the “Aftertouch” Hardware Source on Macro Control B.</p> <p>The amount can be adjusted continuously [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).</p>

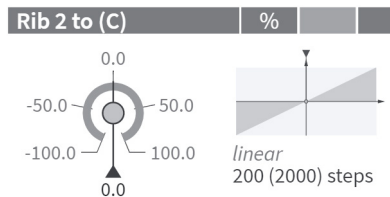
AT to (C) % <input type="checkbox"/> 	<p>Determines the influence of the “Aftertouch” Hardware Source on Macro Control C.</p> <p>The amount can be adjusted continuously [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).</p>
AT to (D) % <input type="checkbox"/> 	<p>Determines the influence of the “Aftertouch” Hardware Source on Macro Control D.</p> <p>The amount can be adjusted continuously [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).</p>
Rib 1 to (A) % <input type="checkbox"/> 	<p>Determines the influence of the “Ribbon 1” Hardware Source on Macro Control A. If the Ribbon is a returning Hardware Source, the amount can be adjusted continuously, otherwise it will be switch-like. Continuous amounts can be set [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).</p>
Rib 1 to (B) % <input type="checkbox"/> 	<p>Determines the influence of the “Ribbon 1” Hardware Source on Macro Control B. If the Ribbon is a returning Hardware Source, the amount can be adjusted continuously, otherwise it will be switch-like. Continuous amounts can be set [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).</p>
Rib 1 to (C) % <input type="checkbox"/> 	<p>Determines the influence of the “Ribbon 1” Hardware Source on Macro Control C. If the Ribbon is a returning Hardware Source, the amount can be adjusted continuously, otherwise it will be switch-like. Continuous amounts can be set [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).</p>
Rib 1 to (D) % <input type="checkbox"/> 	<p>Determines the influence of the “Ribbon 1” Hardware Source on Macro Control D. If the Ribbon is a returning Hardware Source, the amount can be adjusted continuously, otherwise it will be switch-like. Continuous amounts can be set [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).</p>



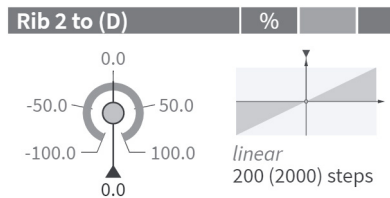
Determines the influence of the “Ribbon 2” Hardware Source on Macro Control A. If the Ribbon is a returning Hardware Source, the amount can be adjusted continuously, otherwise it will be switch-like. Continuous amounts can be set [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).



Determines the influence of the “Ribbon 2” Hardware Source on Macro Control B. If the Ribbon is a returning Hardware Source, the amount can be adjusted continuously, otherwise it will be switch-like. Continuous amounts can be set [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).

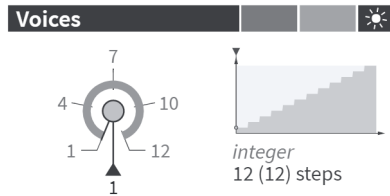


Determines the influence of the “Ribbon 2” Hardware Source on Macro Control C. If the Ribbon is a returning Hardware Source, the amount can be adjusted continuously, otherwise it will be switch-like. Continuous amounts can be set [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).

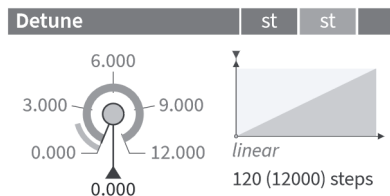


Determines the influence of the “Ribbon 2” Hardware Source on Macro Control D. If the Ribbon is a returning Hardware Source, the amount can be adjusted continuously, otherwise it will be switch-like. Continuous amounts can be set [in percent] and can cover the whole range of the Macro Control in both directions (-100.0 % ... 100.0 %).

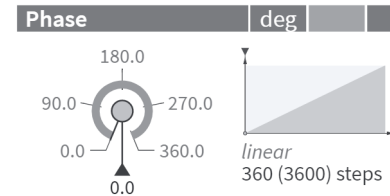
Unison



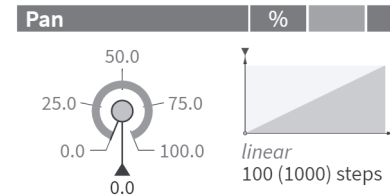
Number of unison voices that will be assigned to a key. At 1, the unison effect is disabled.



Spreading of the pitches of the unison voices [in semitones]. The Fine mode allows for adjustments in steps of 0.001 semitones (0.1 cents).

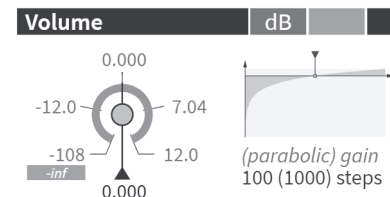


Spreading the (Oscillators’) start phases of the unison voices to get different phase cancellations at the beginning of a note. The value is the phase range [in degrees] covered by a group of unison voices.

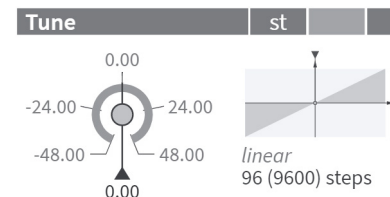


Voice panning inside of a group of unison voices. At 100.0 %, the voices are spread over the full stereo base.

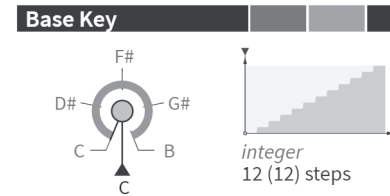
Master / Scale Group



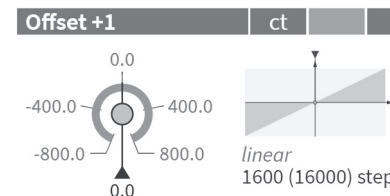
Master volume [in dB], applied at the end of the effect chain, before the soft clipper.



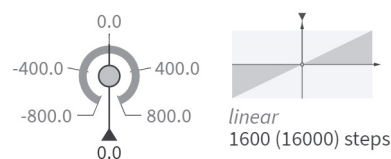
Global pitch transposition [in semitones].



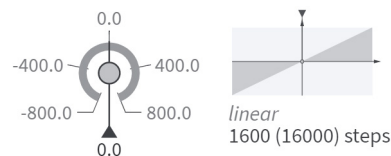
Sets the base key for the custom scale. The scale is defined for the eleven keys above the base key and will be applied to all octaves accordingly.



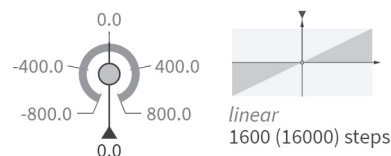
Offset of the first key following the base key [in cents]. At zero, the interval to the base key would be the minor second of the equally tempered scale.

Offset +2 ct

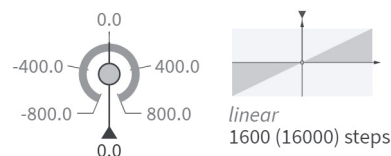
Offset of the 2nd key following the base key [in cents]. At zero, the interval to the base key would be the major second of the equally tempered scale.

Offset +3 ct

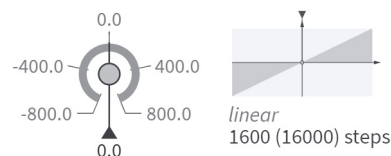
Offset of the 3rd key following the base key [in cents]. At zero, the interval to the base key would be the minor third of the equally tempered scale.

Offset +4 ct

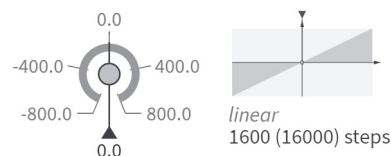
Offset of the 4th key following the base key [in cents]. At zero, the interval to the base key would be the major third of the equally tempered scale.

Offset +5 ct

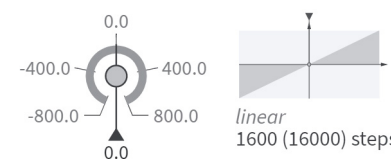
Offset of the 5th key following the base key [in cents]. At zero, the interval to the base key would be the quarter of the equally tempered scale.

Offset +6 ct

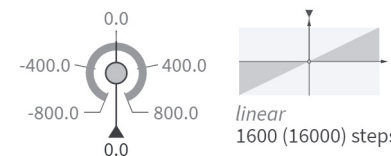
Offset of the 6th key following the base key [in cents]. At zero, the interval to the base key would be the tritone of the equally tempered scale.

Offset +7 ct

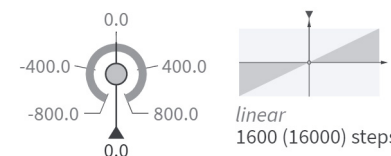
Offset of the 7th key following the base key [in cents]. At zero, the interval to the base key would be the fifth of the equally tempered scale.

Offset +8 ct

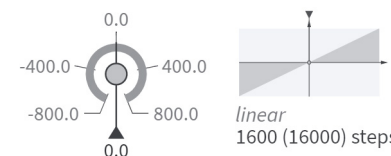
Offset of the 8th key following the base key [in cents]. At zero, the interval to the base key would be the minor sixth of the equally tempered scale.

Offset +9 ct

Offset of the 9th key following the base key [in cents]. At zero, the interval to the base key would be the major sixth of the equally tempered scale.

Offset +10 ct

Offset of the 10th key following the base key [in cents]. At zero, the interval to the base key would be the minor seventh of the equally tempered scale.

Offset +11 ct

Offset of the 11th key following the base key [in cents]. At zero, the interval to the base key would be the major seventh of the equally tempered scale.