Eventide®

Eclipse™

Algorithms Manual

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Eclipse Algorithms Manual – Contents

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Introduction

Some perusal of this manual may aid in general understanding, and assist in 'tweaking' or customizing the factory presets – however, in-depth study of the operation of the algorithms is only necessary for those who wish to get the most from the unit by creating their own presets.

The rest of this manual assumes some understanding of the structure and components of signal processing based effects, but such understanding is not otherwise necessary for normal operation of the Eclipse. People who tend mainly to load and tweak the factory effects can skip the rest of this manual if they so wish.

Each Eclipse preset is made up of one or two *algorithms*. The word *algorithm* can be defined as a "way of doing something." In Eclipse, it is a way to produce an effect – a block of signal processing elements, such as mixers, shifters, compressors, etc. Each of the algorithms can be a powerful and sometimes complex effect in itself.

The factory presets numbered 100-199 should be viewed as examples of the algorithms, and can be used as sources of the 'un-tweaked' algorithm for preset creation. In some cases their names are subtly different from those of the underlying algorithms – this is usually for display or readability purposes. The corresponding preset for each algorithm is given in parentheses below the title e.g. *(6 Chorusdelays)*

Some algorithms have very many parameters, many of which rarely need to be changed in normal use. For this reason, the most important parameters are usually brought out to *hotkeys* to allow greater accessibility. The assignment and naming of *hotkeys* is covered in the **Operators Manual**.

General details applying to many algorithms

- With four voice stereo input delay and shifter effects, voices #1 and #3 are fed from the left input (channel #1), and voices #2 and #4 are fed from the right input (channel #2).
- All *LFO*s (Low Frequency Oscillators) may be re-triggered (synched) by an external audio source, by the front panel <TAP> button, MIDI 'start' commands or by MIDI BPM. These sources may be enabled from the modulation block (retrig) menu.
- all <t_parameters> signify *tap tempo* based rhythmic note value selection (dot 1/4, whole note, etc.) Hz, Sec, mS, etc. These are referenced from the global tempo, the 'modulation block' (taptime) parameters and external MIDIclock. Note that a 1/4 note is defined as the time between averaged <TAP> button hits.
- Many effects have a 'send' parameter - this is a link to automate the input signal remotely and can be set up as a 'send/volume pedal'.
- Unless noted otherwise, most delay times (delay and shifters) can be set to up to two seconds.
- Any pitch shift effect should be calibrated to the source signal. These effects can be program-material dependent, so some tweaking will often give better results, especially diatonic shifters, which usually need the key signature and scale to be specified.
- Any envelope/peak detector effect (including fm modulation effects) should be calibrated to the source material for the best results. This is usually a matter of adjusting for level and dynamics of the input signal.
- Any ADSR effect should be calibrated to the source material for best results. This is usually a matter of matching the dynamics of the input signal.
- Many algorithms have an "S" somewhere in their signal path. That's the point at which the (SEND) or (MANUAL_P) level control operates.

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delays

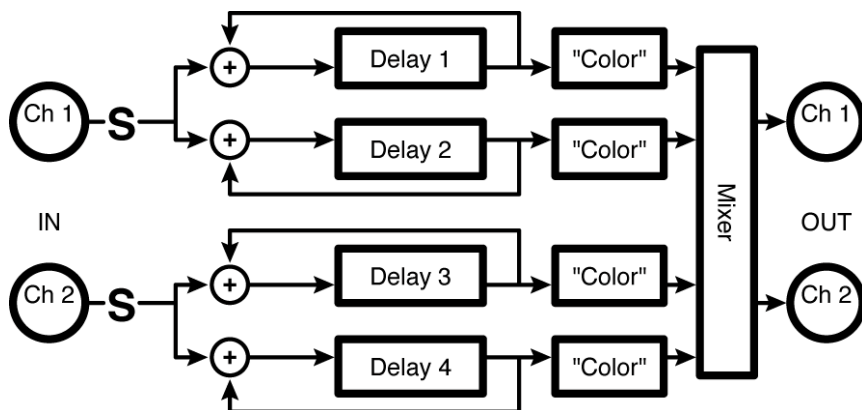
m_banddelays

(10 4   Banddelays)

Four parallel delays with independent feedback paths feeding resonant bandpass filters. Tempo can be used to control delay time.
Stereo in, stereo out

*Parameters***:**

Masters	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
m_Level	*Scales all the individual delay levels, effectively serving as a master volume.*
m_Delay	*Scales all the individual delay times, effectively serving as a master delay (e.g. if the four delays are set to 200ms, 150ms, 100ms, and 50ms, setting M_Delay to 50% cuts those times in half (100ms, 75ms, 50ms, and 25ms)).*
m_Freq	*Scales all the individual band frequencies, effectively serving as a master frequency.*
m_Q	*Scales all the individual band Q's, effectively serving as a master Q. High values give high resonance, low values give low resonance.*
m_Fback	*Scales all the individual delay feedbacks, effectively serving as a master feedback (i.e. if M_Fback is set to 0%, all the delays will have no feedback, regardless of their individual settings. Alternatively, you could also have no feedback with M_Fback set to 100% if you set each delay's feedback parameter to 0%.).*
m_Pan	*Scales all the individual delay pans, effectively serving as an image spread control (i.e. if M_Pan is set to 0%, all the delays will be panned center, regardless of their individual settings).*

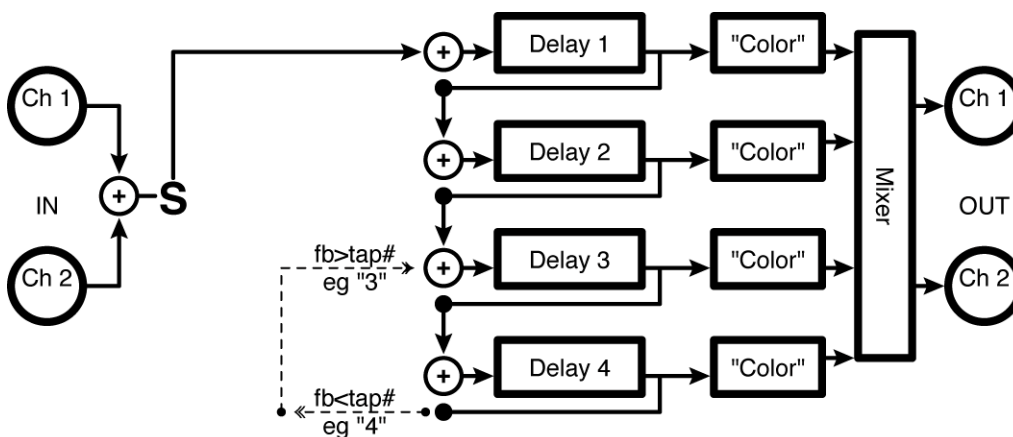| Delay#^(x = 1, 2, 3, or 4)	
Level x	*Controls the level of delay x.*
Delay x	*Either reflects the physical delay time as entered by T_Delay x (given the current tempo) or allows you to enter the delay time in milliseconds.*
T_Delay x	*Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay control "off".*
Freqx	*Controls the frequency that delay x's bandpass filter ("Color") will center on.*

Qx	*Controls how sharp (resonant) said bandpass filter will be.*
Fback	*Controls how much of delay x's output is reapplied to its input.*
Panx	*Controls where in the stereo field delay x will appear (negative values scoot it left and positive values scoot it right).*

m_bandtaps

(105   Bandtaps)

Four series-connected delays with bandpass filters feeding a stereo mixer. A wide range of feedback options allows rhythmically complex and colorful patterns to be quickly and easily created. Tempo may be used to control delay times.

Summed in, stereo out

Parameters:

Masters	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
m_Level	*Scales all the individual delay levels, effectively serving as a master volume.*
m_Delay	*Scales all the individual delay times, effectively serving as a master delay (e.g. if the four delays are set to 200ms, 150ms, 100ms, and 50ms, setting M_Delay to 50% cuts those times in half (100ms, 75ms, 50ms, and 25ms)).*
m_Freq	*Scales all the individual band frequencies, effectively serving as a master frequency.*
m_Q	*Scales all the individual band Q's, effectively serving as a master Q. High values give high resonance, low values give low resonance.*
Fback	*Scales all the individual delay feedbacks, effectively serving as a master feedback (i.e. if Fback is set to 0%, all the delays will have no feedback, regardless of their individual settings. Alternatively, you could also have no feedback with M_Fback set to 100% if you set each delay's feedback parameter to 0%.).*
Fb<tap#	*Selects which of the four delays will provide the output for feedback (e.g. as shown in the diagram, delay 4 has been selected).*
Fb>tap#	*Selects where the feedback output should be "re-injected" (e.g. as shown in the diagram, the feedback output will be "re-injected" into the input of delay 3).*
m_Pan	*Scales all the individual delay pans, effectively serving as an image spread control (i.e. if M_Pan is set to 0%, all the delays will be panned center, regardless of their individual settings).*

Delays^(x = 1, 2, 3, or 4)

Tapx	Either reflects the physical delay time of tap x as entered by T_Tap x (given the current tempo) or allows you to enter the delay time in milliseconds.
T_Tap x	Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay control "off".
Level x	Controls the level of delay x.
Pan x	Controls where in the stereo field the delay x output will appear. Negative values scoot it left and positive values scoot it right.

Filters^(x = 1, 2, 3, or 4)

Freq x	Controls the frequency that delay x's bandpass filter ("Color") will center on.
Q x	Controls how sharp (resonant) said bandpass filter will be.

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m_chorusdelays

(106   Chorusdelays)

Four parallel delays with independent feedback paths feed choruses. Tempo can be used to control delay time and modulation rate.
Stereo in, stereo out

See "m_ banddelays" above for diagram.

Masters

Send	Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
m_Level	Scales all the individual delay levels, effectively serving as a master volume.
m_Delay	Scales all the individual delay times, effectively serving as a master delay (e.g. if the four delays are set to 200ms, 150ms, 100ms, and 50ms, setting M_Delay to 50% cuts those times in half (100ms, 75ms, 50ms, and 25ms)).
m_Rate	Scales all the individual delay sweep rates, effectively serving as a master rate.
m_Depth	Scales all the individual delay sweep depths, effectively serving as a master depth.
m_Fback	Scales all the individual delay feedbacks, effectively serving as a master feedback (i.e. if M_Fback is set to 0%, all the delays we have no feedback, regardless of their individual settings. Alternatively, you could also have no feedback with M_Fback set to 100% if you set each delay's feedback parameter to 0%.).
m_Pan	Scales all the individual delay pans, effectively serving as an image spread control (i.e. if M_Pan is set to 0%, all the delays will be panned center, regardless of their individual settings).
Retrig	Whenever the sweep rate is changed (either manually or by updating the tempo), the new rate is crossfaded with the old rate. The value of Retrig controls how long the crossfade is as a percentage of the old rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old rate.
Angle	Given most configurations of the Eclipse, the sweep rate will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the sweep begins.

Delay#1   (similarly 2, 3, and 4)

Level1	Controls the level of delay 1.
Delay1	Either reflects the physical delay time as entered by T_Delay1 (given the current tempo) or allows you to enter the delay time in milliseconds.
T_Delay1	Controls delay 1 in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay control "off".
Rate1	Controls the sweep rate for delay 1 if T_Rate1 is set to "off" or reflects the physical sweep rate as selected by T_Rate1 and the system tempo.

T_Rate1   Controls the sweep rate of delay 1 in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic sweep generation "off".

Depth1   Controls the sweep depth of delay 1.

Fback1   Controls how much of delay 1's output is reapplied to its input.

Pan1   Controls where in the stereo field delay 1 will appear (negative values scoot it left and positive values scoot it right).

m_chorustaps

(107   Chorustaps)

Four series-connected delays with choruses feeding a stereo mixer. A wide range of feedback options allows rhythmically complex and colorful patterns to be quickly and easily created. Tempo may be used to control delay times and modulation rates

Summed in, stereo out

See "m_ bandtaps" above for diagram

Masters	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
m_Level	Scales all the individual delay levels, effectively serving as a master volume.
m_Delay	Scales all the individual delay times, effectively serving as a master delay (e.g. if the four delays are set to 200ms, 150ms, 100ms, and 50ms, setting M_Delay to 50% cuts those times in half (100ms, 75ms, 50ms, and 25ms)).
m_Rate	Scales all the individual delay sweep rates, effectively serving as a master rate.
m_Depth	Scales all the individual delay sweep depths, effectively serving as a master depth.
Fback	Scales all the individual delay feedbacks, effectively serving as a master feedback (i.e. if Fback is set to 0%, all the delays will have no feedback, regardless of their individual settings. Alternatively, you could also have no feedback with M_Fback set to 100% if you set each delay's feedback parameter to 0%.).
Fb<tap#	Selects which of the four delays will provide the output for feedback (e.g. as shown in the diagram, delay 4 has been selected).
Fb>tap#	Selects where the feedback output should be "reinjected" (e.g. as shown in the diagram, the feedback output will be "reinjected" into the input of delay 3).
m_Pan	Scales all the individual delay pans, effectively serving as an image spread control (i.e. if M_Pan is set to 0%, all the delays will be panned center, regardless of their individual settings).
Retrig	Whenever the sweep rate is changed (either manually or by updating the tempo), the new rate is crossfaded with the old rate. The value of Retrig controls how long the crossfade is as a percentage of the old rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old rate.
Angle	Given most configurations of the Eclipse, the sweep rate will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the sweep begins.

Delays^(x = 1, 2, 3, or 4)	
Tapx	Either reflects the physical delay time of tap x as entered by T_Tap x (given the current tempo) or allows you to enter the delay time in milliseconds.
T_Tap x	Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".
Level x	Controls the level of delay x.
Pan x	Controls where in the stereo field delay x will appear (negative values scoot it left and positive values scoot it right).

Eclipse Algorithms Manual

LFOs^(x = 1, 2, 3, or 4)

Rate x	Controls the sweep rate for delay x if T_Ratex is set to "off" or reflects the physical sweep rate as selected by T_Ratex and the system tempo.
T_Rate x	Controls the sweep rate of delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic sweep generation "off".
Depth x	Controls the sweep depth of delay x.

m_combdelays

(108   Combdelays)

Four parallel delays with independent feedback paths feed resonators. Tempo can be used to control delay time.

Stereo in, stereo out

See "m_ banddelays" above for diagram.

Masters

Send	Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
m_Level	Scales all the individual delay levels, effectively serving as a master volume.
m_Delay	Scales all the individual delay times, effectively serving as a master delay (e.g. if the four delays are set to 200ms, 150ms, 100ms, and 50ms, setting M_Delay to 50% cuts those times in half (100ms, 75ms, 50ms, and 25ms)).
m_Cmix	Scales all the individual delay "comb-to-dry" ratios, effectively serving as a master "comb-to-dry" ratio.
m_Comb	Scales all the individual delay comb depths, effectively serving as a master comb depth. Changing the comb depth changes the timbre of the effect.
m_Decay	Scales all the individual delay comb "resonances", effectively serving as a master decay.
m_Fback	Scales all the individual delay feedbacks, effectively serving as a master feedback (i.e. if M_Fback is set to 0%, all the delays will have no feedback, regardless of their individual settings. Alternatively, you could also have no feedback with M_Fback set to 100% if you set each delay's feedback parameter to 0%.).
m_Pan	Scales all the individual delay pans, effectively serving as an image spread control (i.e. if M_Pan is set to 0%, all the delays will be panned center, regardless of their individual settings).

Delay#1

Level1	Controls the level of delay 1.
Delay1	Either reflects the physical delay time as entered by T_Delay1 (given the current tempo) or allows you to enter the delay time in milliseconds.
T_Delay1	Controls delay 1 in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".
CMix1	Controls the "comb-to-dry" ratio for delay line 1.
Comb1	Controls the comb depth for delay line 1; adjusts the timbre of the effect.
Decay1	Controls the comb "resonance" for delay line 1.
Fback	Controls how much of delay 1's output is reapplied to its input.
Pan1	Controls where in the stereo field delay 1 will appear (negative values scoot it left and positive values scoot it right).

m_combtaps

(109   Combtaps)

Four series-connected delays with resonators feeding a stereo mixer. A wide range of feedback options allows rhythmically complex and colorful patterns to be quickly and easily created. Tempo may be used to control delay times.

Summed in, stereo out

See "m_ bandtaps" above for diagram.

Masters	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
m_Level	Scales all the individual delay levels, effectively serving as a master volume.
m_Delay	Scales all the individual delay times, effectively serving as a master delay (e.g. if the four delays are set to 200ms, 150ms, 100ms, and 50ms, setting M_Delay to 50% cuts those times in half (100ms, 75ms, 50ms, and 25ms)).
m_Cmix	Scales all the individual delay "comb-to-dry" ratios, effectively serving as a master "comb-to-dry" ratio.
m_Comb	Scales all the individual delay comb depths, effectively serving as a master comb depth. Changing the comb depth changes the timbre of the effect.
m_Decay	Scales all the individual delay comb "resonances", effectively serving as a master decay.
Fback	Scales all the individual delay feedbacks, effectively serving as a master feedback (i.e. if Fback is set to 0%, all the delays will have no feedback, regardless of their individual settings. Alternatively, you could also have no feedback with M_Fback set to 100% if you set each delay's feedback parameter to 0%.).
Fb<tap#	Selects which of the four delays will provide the output for feedback (e.g. as shown in the diagram, delay 4 has been selected).
Fb>tap#	Selects where the feedback output should be "reinjected" (e.g. as shown in the diagram, the feedback output will be "reinjected" into the input of delay 3).
m_Pan	Scales all the individual delay pans, effectively serving as an image spread control (i.e. if M_Pan is set to 0%, all the delays will be panned center, regardless of their individual settings).

Delays^(x = 1, 2, 3, or 4)	
Tap x	Either reflects the physical delay time of tap x as entered by T_Tap x (given the current tempo) or allows you to enter the delay time in milliseconds.
T_Tap x	Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".
Level x	Controls the level of delay x.
Pan x	Controls where in the stereo field delay x will appear (negative values scoot it left and positive values scoot it right).

Combs^(x = 1, 2, 3, or 4)	
Cmix x	Controls the "comb-to-dry" ratio for delay line x.
Comb x	Controls the comb depth for delay line x; adjusts the timbre of the effect.
Decay x	Controls the comb "resonance" for delay line x.

m_ringdelays

(112   Ringdelays)

Four parallel delays with independent feedback paths feed frequency shifters (single sideband ring modulation). Tempo may be used to control delay times and ringing.
Stereo in, stereo out

See "m_ banddelays" above for diagram.

Masters	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
m_Level	*Scales all the individual delay levels, effectively serving as a master volume.*
m_Delay	*Scales all the individual delay times, effectively serving as a master delay (e.g. if the four delays are set to 200ms, 150ms, 100ms, and 50ms, setting M_Delay to 50% cuts those times in half (100ms, 75ms, 50ms, and 25ms)).*
m_Fback	*Scales all the individual delay feedbacks, effectively serving as a master feedback (i.e. if M_Fback is set to 0%, all the delays will have no feedback, regardless of their individual settings Alternatively, you could also have no feedback with M_Fback set to 100% if you set each delay's feedback parameter to 0%.).*
m_Pan	*Scales all the individual delay pans, effectively serving as an image spread control (i.e. if M_Pan is set to 0%, all the delays will be panned center, regardless of their individual settings).*
m_Rmix	*Scales all the individual delay "ring-to-dry" ratios, effectively serving as a master "ring-to-dry" ratio.*
m_Ring	*Scales all the individual delay ring depths, effectively serving as a master ring depth. Changing the ring depth changes the timbre of the effect.*

Delay#1	
Level1	*Controls the level of delay 1.*
Delay1	*Either reflects the physical delay time as entered by T_Delay1 (given the current tempo) or allows you to enter the delay time in milliseconds.*
T_Delay1	*Controls delay 1 in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".*
RMix1	*Controls the "ring-to-dry" ratio for delay line 1.*
Ring1	*Either reflects the physical ring depth as entered by T_Ring1 (given the current tempo) or allows you to enter the ring depth in Hz.*
T_Ring1	*Controls ring depth 1 in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic ring depth generation "off".*
Fback	*Controls how much of delay 1's output is reapplied to its input.*
Pan1	*Controls where in the stereo field delay 1 will appear (negative values scoot it left and positive values scoot it right).*

m_ringtaps

(113   Ringtaps)

Four series-connected delays with frequency shifters feeding a stereo mixer. A wide range of feedback options allows rhythmically complex and colorful patterns to be quickly and easily created. Tempo may be used to control delay times and ringing.

Summed in, stereo out

See "m_bandtaps" above for diagram.

Masters	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
m_Level	Scales all the individual delay levels, effectively serving as a master volume.
m_Delay	Scales all the individual delay times, effectively serving as a master delay (e.g. if the four delays are set to 200ms, 150ms, 100ms, and 50ms, setting M_Delay to 50% cuts those times in half (100ms, 75ms, 50ms, and 25ms)).
m_Rmix	Scales all the individual delay "ring-to-dry" ratios, effectively serving as a master "ring-to-dry" ratio.
m_Ring	Scales all the individual delay ring depths, effectively serving as a master ring depth. Changing the ring depth changes the timbre of the effect.
Fback	Scales all the individual delay feedbacks, effectively serving as a master feedback (i.e. if Fback is set to 0%, all the delays will have no feedback, regardless of their individual settings. Alternatively, you could also have no feedback with M_Fback set to 100% if you set each delay's feedback parameter to 0%.).
Fb<tap#	Selects which of the four delays will provide the output for feedback (e.g. as shown in the diagram, delay 4 has been selected).
Fb>tap#	Selects where the feedback output should be "reinjected" (e.g. as shown in the diagram, the feedback output will be "reinjected" into the input of delay 3).
m_Pan	Scales all the individual delay pans, effectively serving as an image spread control (i.e. if M_Pan is set to 0%, all the delays will be panned center, regardless of their individual settings).

Delays^(x = 1, 2, 3, or 4)	
Tap x	Either reflects the physical delay time of tap x as entered by T_Tap x (given the current tempo) or allows you to enter the delay time in milliseconds.
T_Tap x	Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".
Level x	Controls the level of delay x.
Pan x	Controls where in the stereo field the delay x output will appear (negative values scoot it left and positive values scoot it right).

Rings^(x = 1, 2, 3, or 4)	
Rmix x	Controls the "ring-to-dry" ratio for delay line 1.
Ring1	Either reflects the physical ring depth as entered by T_Ringx (given the current tempo) or allows you to enter the ring depth in Hz.
T_Ring x	Controls ring depth x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic ring depth generation "off".

m_pandelays

(114   Pandelays)

Four parallel delays with independent feedback paths may be panned across the stereo output. Tempo may be used to control delay times and pan sweep rates.
Stereo in, stereo out

See "m_ banddelays" above for diagram and parameters.

Masters	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
m_Level	*Scales all the individual delay levels, effectively serving as a master volume.*
m_Delay	*Scales all the individual delay times, effectively serving as a master delay (e.g. if the four delays are set to 200ms, 150ms, 100ms, and 50ms, setting M_Delay to 50% cuts those times in half (100ms, 75ms, 50ms, and 25ms)).*
m_Fback	*Scales all the individual delay feedbacks, effectively serving as a master feedback (i.e. if M_Fback is set to 0%, all the delays will have no feedback, regardless of their individual settings. Alternatively, you could also have no feedback with M_Fback set to 100% if you set each delay's feedback parameter to 0%.).*
m_Rate	*Scales all the individual delay pan rates, effectively serving as a master rate.*
m_Depth	*Scales all the individual delay pan depths, effectively serving as a master depth.*
Retrig	*Whenever the pan rate is changed (either manually or by updating the tempo), the new rate is cross-faded with the old rate. The value of Retrig controls how long the cros-sfade is as a percentage of the old rate. Thus, a value of 100% means that the cross-fade will be complete in one cycle of the old rate.*
Angle	*Given most configurations of the Eclipse, the pan rate will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the sweep begins.*

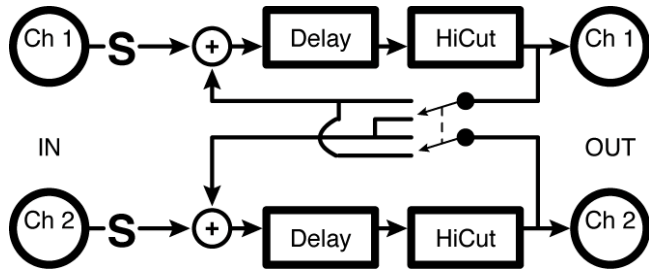| *Delay#1* *(similarily 2,3,4)*	
Level1	*Controls the level of delay 1.*
Delay1	*Either reflects the physical delay time as entered by T_Delay1 (given the current tempo) or allows you to enter the delay time in milliseconds.*
T_Delay1	*Controls delay 1 in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".*
Fback	*Controls how much of delay 1's output is reapplied to its input.*
Rate1	*Controls the pan rate for delay 1 if T_Rate1 is set to "off" or reflects the physical pan rate as selected by T_Rate1 and the system tempo.*
T_Rate1	*Controls the pan rate of delay 1 in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic pan generation "off".*
Depth1	*Controls the pan depth of delay 1.*

st chorus delays

(115   St Chorus)

Dual modulating delays with various feedback options are followed by a hi-cut filter. Tempo may be used to control the delay's sweep rate. Each delay may be set at up to two seconds, allowing longer delay times than similar effects. The left and right channels sweep in opposition.
Stereo in, stereo out

Delays	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
Level	*Controls the output levels of both delays.*
Delay1	*Controls the delay time of delay 1 in milliseconds (no tempo function is available - this is a chorus effect, after all).*
Delay2	*Controls the delay time of delay 2 in milliseconds.*
Depth1	*Controls the depth of modulation for delay 1.*
Depth2	*Controls the depth of modulation for delay 2.*
Hicut	*Controls the lowpass filter cutoff frequency.*
Fback	*Controls how much of each delay's output gets reapplied to its own input or to the other delay's input.*
FB Type	*Determines if the feedback from each delay will return to its own input or to the other delay's input.*
Image	*Controls how wide the stereo chorus image is.*

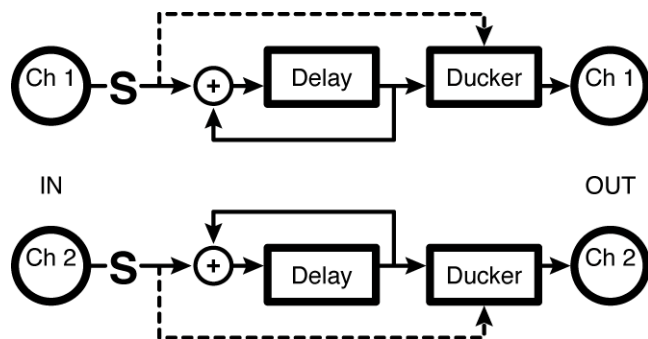| LFO	
Rate	*Controls the modulation rate if T_Rate1 is set to "off" or reflects the physical modulation rate as selected by T_Rate and the system tempo.*
T_Rate	*Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation control "off".*
Shape	*Determines the shape of the modulating signal.*
Duty	*Controls the "duty cycle" of the modulating wave for any waveform other than "sine". For example, if you have selected "triangle" under Shape and set Duty to "0%", the triangle will morph to a sawtooth (very fast rise, slow fall). With Duty at "100%", the triangle will morph to a ramp (slow rise, very fast fall).*
Polarity	*When set to "unipolar", modulations add the value of Depth to Delay. When set to "bipolar", modulations add the value of Depth to Delay and subtract the value of Depth from Delay.*
Retrig	*Whenever **Rate** is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.*
Angle	*Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.*

ducked delays

(116   DuckedDelays)

Two delay lines followed by a stereo compressor whose side-chain is fed by a sum of chan#1 and #2.
Volume ducks/compresses delays out of the way. Be careful with compressor gain! Tempo may be used
to control delay times.
Stereo in, stereo out

Delay	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
Level	Controls the output levels of both delays.
Delay1	Either reflects the physical delay time as entered by T_Delay1 (given the current tempo) or allows you to enter the delay time in milliseconds.
Delay2	Either reflects the physical delay time as entered by T_Delay2 (given the current tempo) or allows you to enter the delay time in milliseconds.
T_Delay1	Controls delay 1 in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".
T_Delay2	Controls delay 2 in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".
Fback1	Controls how much of delay 1's output is reapplied to its input.
Fback2	Controls how much of delay 2's output is reapplied to its input.
Image	Controls how wide the delay image is.

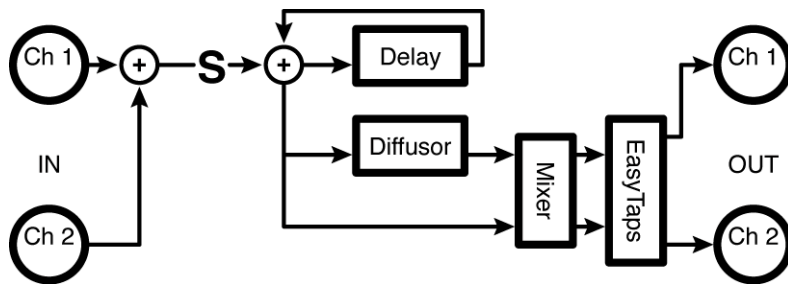| Ducking	
Thresh	Sets the input threshold above which compression of the delay signal takes place and below which the delay signal is left alone.
Ratio	Controls the input to output ratio for signals above the threshold. For example, "6:1" produces 1dB of output for every 6dB of input.
Gain	Adjusts the output level to improve gain structure.
Attack	Controls how quickly the compressor "clamps down" on a signal after it exceeds the threshold.
Decay	Controls how quickly the compressor "lets go" of a signal once it falls below the threshold.
GR 1	Displays how much gain reduction (compression) is taking place after delay 1.
GR 2	Displays how much gain reduction (compression) is taking place after delay 2.

ultratap 2

(117   Ultratap 2)

This extension of a popular Eventide H3000 structure contains four serial two-second diffusors feeding a 36 tap, two-second *multitap* delay (*easytaps* below) and a tempo-controlled two-second feedback delay. It is great for pre-reverb or strange delay/reverberant effects.
Summed in, stereo out

Diffusor^(x = 1, 2, 3, 4)	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
Dmix	Controls the amount of diffusion.
Coef x	Controls the amount of feedback for diffusor x, which determines the "fuzziness" or "distinctness" of the diffusor.
Diffusion	A master feedback control for all the diffusors.
DSize	Controls the diffusor's "size" by scaling its delay times.
Delay x	Controls the total delay time for diffusor x.

Multitap	
Level	Controls the output level of the multitap delay.
FBDelay	Controls the delay time of the feedback delay.
T_FBDela	Controls the feedback delay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".
Fback	The amount of feedback applied to the feedback delay.
# Taps	Specifies the number of delay taps that the algorithm will use.
Length	Controls the total delay length.
Random	Controls the degree to which the Eclipse randomizes tap times to reduce resonance.
DlyShape	Controls the delay "shape" or tap length as a function of tap number.
AmpShape	Controls the amplitude "shape" or the level of each of tap as a function of tap number.
PanShape	Controls the pan "shape" or the panning of each tap as a function of tap number.
AAlpha	Controls the exponential shape of the amplitude only when AmpShape is set to "exp_inc" or "exp_dec".
DAlpha	Controls the exponential shape of the tap times only when DlyShape is set to "exp_inc" or "exp_dec".

mono loop (20)

(119   Loop20)

This simple effect contains a mono 20-second delay with hi-cut filter and pan. Tempo may be used to control delay times.

Summed in, stereo out

Loops

Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the input to the loop with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
Level	*Controls the output level of the loop.*
Loop	*Either reflects the physical loop time as entered by T_Loop (given the current tempo) or allows you to enter the loop time in milliseconds.*
T_Loop	*Controls the length of the loop in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic loop generation "off".*
Hicut	*Controls the lowpass filter cutoff frequency.*
Pan	*Pans the loop in the stereo field.*
Loop	*Monitors the input to the loop*

mono loop (10)

(118   Loop10)

This version of "mono loop (20)" is suitable for use at 96kHz sampling.

Summed in, stereo out

See "mono loop (20)" above for diagram and parameters.

dual loops (10)

(121   DualLoop10)

A true stereo version of "mono loop (20)."
Stereo in, stereo out

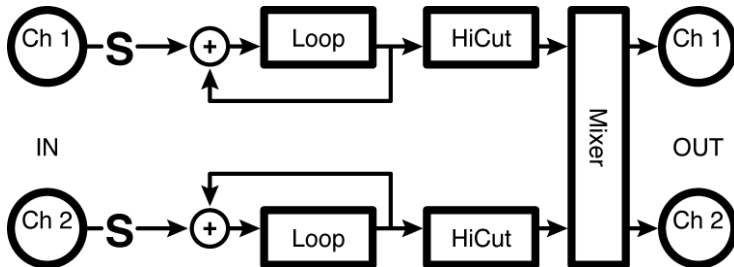| Master	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the input to the loop with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
Level	*Controls the output level of both loops.*
Loop	*Either reflects the physical loop time of both loops as entered by T_Loop (given the current tempo) or allows you to enter the loop time in milliseconds.*
T_Loop	*Controls the length of both loops in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic loop generation "off".*
Fback	*Controls how much of each loop's output is reapplied to its input or the other loop's input.*
FB Type	*Determines if the feedback from each loop will return to its own input or to the other loop's input.*
Hicut	*Controls the lowpass filter cutoff frequency.*
Image	*Controls how wide the loop image is.*
Loop1	*Monitors the input to loop 1.*
Loop2	*Monitors the input to loop 2.*

Loop#1
Parameters similar to those described under "Mono Loop" above, dedicated to Loop 1; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

Loop#2
Parameters similar to those described under "Mono Loop" above, dedicated to Loop 2; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

dual loops (5)

(120   DualLoop5)

This version of "dual loops (10)" is suitable for use at 96kHz sampling.
Stereo in, stereo out

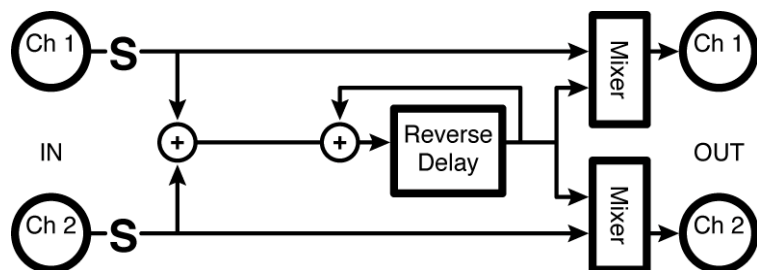See "dual loops (10)" above for diagram and parameters.

mono reverse (20)

(122   Reverse10)

This algorithm is a mono 20-second *reverse delay*. A *reverse delay* is unique in that the <tap> button or other trigger source can both set the delay time and trigger playback from the unit.. This means that the first button press starts filling the delay, while the second press starts backwards playback. Think of it as a reverse sampler or looper. Tempo may be used to control delay times.
Summed in, stereo out

Reverse

Send   Adjusts the level at 'S' in the diagram. If you want to modulate the input to the loop with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.

Mix   Controls the wet-to-dry ratio. If set to "0%", you won't hear any effect, while if set to "100%" you'll hear nothing *but* effect.

Delay   Either reflects the physical reverse time as entered by T_Delay (given the current tempo) or allows you to enter the reverse time in milliseconds.

T_Delay Controls the length of the reverse in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic reverse generation "off".

Regen   Controls the feedback around the Reverse Delay, i.e. how much of its output is reapplied to its input.

XFade   Controls the length of crossfade between successive samples (reversed chunks).

Mute   Sets the delay time below which the reverse output will be muted (prevents tempo changes from creating a ridiculously short reverse loop).

Trigger Causes reversed playback to start immediately.

mono reverse (10)

(123   Reverse20)

This version of "mono reverse (20)" is suitable for use at 96kHz sampling.
Summed in, stereo out

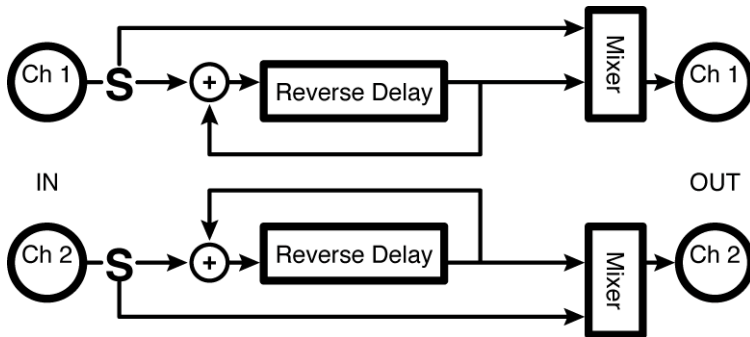See "mono reverse (20)" above for diagram and parameters.

dual reverse (10)

(125   DualReverse10)

This is a 10-second true stereo version of "mono reverse (20)."
Stereo in, stereo out

Master	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the input to the loop with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
Mix	Controls the wet-to-dry ratio. If set to "0%", you won't hear any effect. If set to "100%" you'll hear nothing but effect.
Delay	Either reflects the physical reverse time as entered by T_Delay (given the current tempo) or allows you to enter the reverse time in milliseconds if T_Delay is set to "Off".
T_Delay	Controls the length of the reverse in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic reverse generation "off". The actual reverse time displayed under Delay is a function of this control and the current system tempo.
Fback	Controls how much of the reverse's output is reapplied to its input.
Mute	Sets the delay time below which the reverse output will be muted (prevents tempo changes from creating a ridiculously short reverse loop).
XFade	Controls the length of crossfade between successive samples (reversed chunks).

Rev#1

Parameters similar to those described above, dedicated to Loop 1; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

Rev#2

Parameters similar to those described above, dedicated to Loop 2; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

Trigger	Causes reversed playback to start immediately.

dual reverse (5)

(124   DualReverse5)

This five-second true stereo version of "mono reverse (20)" is suitable for use at 96kHz sampling.
Stereo in, stereo out

See "dual reverse (10)" above for diagram and parameters.

Eclipse Algorithms Manual

dynamics

dual compressors

(126   StereoComp)

A pair of independent "soft knee" compressors featuring 'master' and individual channel parameters as well as metering of gain reduction.
Dual mono in, dual mono out

Master	
Thresh	Sets the input threshold above which compression of the signal takes place and below which the signal is left alone.
S Knee	Controls the width of the "soft knee". The soft knee is a region above the threshold over which the ratio transitions from 1:1 to the selected Ratio. It smoothes out the compression.
Ratio	Controls the input to output ratio for signals above the threshold. For example, "6:1" produces 1dB of output for every 6dB of input.
Gain	Adjusts the output level to improve gain structure.
Attack	Controls how quickly the compressor "clamps down" on a signal after it exceeds the threshold.
Decay	Controls how quickly the compressor "lets go" of a signal once it falls below the threshold.
GR 1	Displays how much gain reduction (compression) is taking place after delay 1.
GR 2	Displays how much gain reduction (compression) is taking place after delay 2.

Comp#1

Parameters similar to those described above, dedicated to compressor 1; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

Comp#2

Parameters similar to those described above, dedicated to compressor 2; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

Eclipse Algorithms Manual

dual noisegates

(127   DualGates)

This is a stereo noise gate with 'master' and individual channel parameters.
Stereo in, stereo out

Master	
Trigg	*Selects which input or inputs trigger the gate to open.*
Thresh	*Sets the input threshold that the trigger must exceed to open the gate.*
Attack	*Controls how quickly the gate opens up once the trigger exceeds the threshold.*
Decay	*Controls how quickly the gate closes once the trigger falls below the threshold.*
Hyster	*Controls the "hysteresis": how far the trigger must fall below the threshold before the gate will open. Higher values prevent spurious triggering.*
GT 1	*Displays the dynamics of gate 1: if the bar is empty, the gate is closed. If the bar is full, the gate is open. Intermediate values represent varying levels of gain reduction.*
GT 2	*Displays the dynamics of gate 2: if the bar is empty, the gate is closed. If the bar is full, the gate is open. Intermediate values represent varying levels of gain reduction.*

Gate#1

Parameters similar to those described above, dedicated to gate 1; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

Gate#2

Parameters similar to those described above, dedicated to gate 2; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")
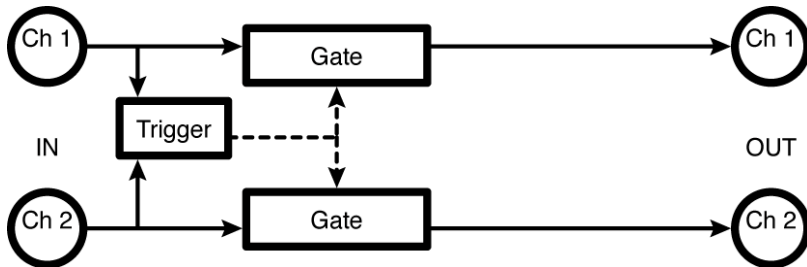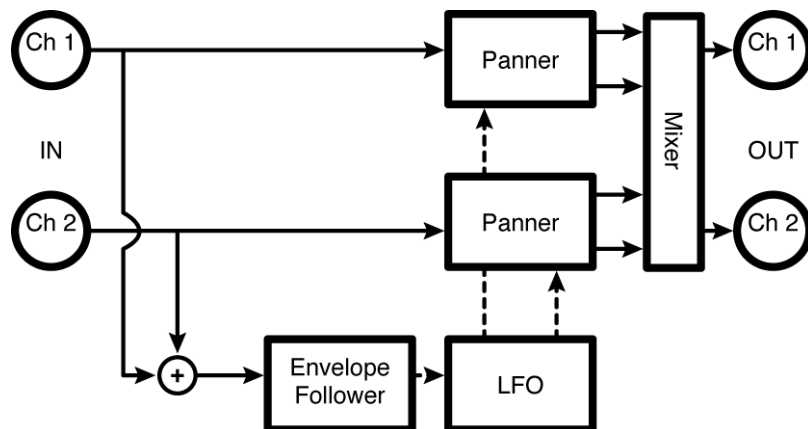
fm panner

(128 FM Panner)

Simple panning with envelope-controlled fm modulation of sweep.
Stereo in, stereo out

FM Pan	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
Depth	Amount of panning, relative to the output stereo field width.
Width	Actual width of output stereo field.
Mode	Determines if the effect will behave like a panner ("sync") or like a tremolo ("oppose").

LFO	
Rate	Controls the modulation rate if T_Rate1 is set to "off".
T_Rate	Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".
FMRate	Analogous to Rate, except that this pertains to modulation of the LFO's signal (not to modulation of the panner directly).
T_FMRate	Analogous to T_Rate, except that this pertains to modulation of the LFO's signal (not to modulation of the panner directly).
Shape	Determines the shape of the modulating signal.
Duty	Controls the "duty cycle" of the modulating wave for any waveform other than "sine". For example, if you have selected "triangle" under Shape and set Duty to "0%", the triangle will morph to a sawtooth (very fast rise, slow fall). With Duty at "100%", the triangle will morph to a ramp (slow rise, very fast fall).
Retrig	Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.
Angle	Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.

Env	
Sens	Controls how sensitive the envelope follower is; alter this parameter to change the "depth" of the FM effect.
Attack	Controls how fast the envelope follower will track an increase in volume.
Decay	Controls how fast the envelope follower will track a decrease in volume.

fm trem

(129   FM Trem)

A simple tremolo with envelope-controlled fm modulation of sweep.
Stereo in, stereo out

FM Trem	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is the place to do it. No zippering allowed.
Mix	Controls the ratio between the dry and wet (tremolo) signal.
Depth	Controls the depth of the modulation. If Mix is at "0%", this effectively controls the volume.
Mode	Determines if the effect will behave like a tremolo ("sync") or like a panner ("oppose").

LFO	
Rate	Controls the modulation rate if T_Rate1 is set to "off" or reflects the physical modulation rate as selected by T_Rate.
T_Rate	Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".
FMRate	Analogous to Rate, except that this pertains to modulation of the LFO's signal (not to modulation of the tremolo directly).
T_FMRate	Analogous to T_Rate, except that this pertains to modulation of the LFO's signal (not to modulation of the tremolo directly).
Shape	Determines the shape of the modulating signal.
Retrig	Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.
Angle	Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.

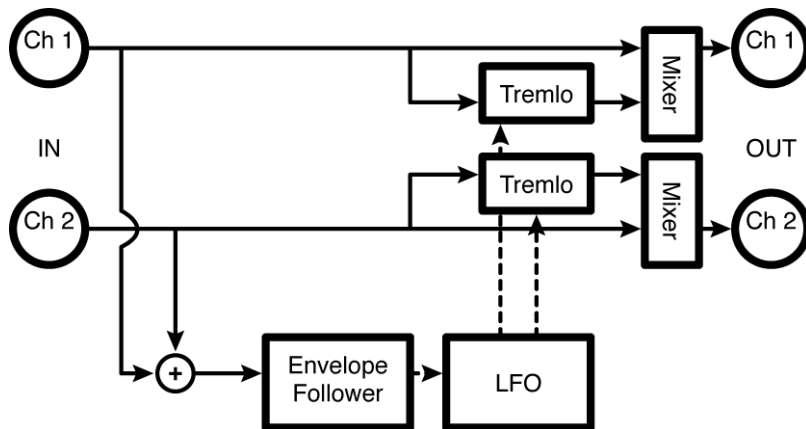| Env	
Sens	Controls how sensitive the envelope follower is; alter this parameter to change the "depth" of the FM effect.
Attack	Controls how fast the envelope follower will track an increase in volume.
Decay	Controls how fast the envelope follower will track a decrease in volume.

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filters

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two band crossover

(130   2BandXover)

This two-band crossover includes pan parameters for the 'hi' and 'low' outputs for each channel.
Stereo in, stereo out

X-Over	
X-Over	*Selects the frequency at which the crossover divides the signal.*
Type	*Selects the "windowing" type. In critical applications, A/B for best results.*

Mixer	
Lows1	*Controls the level of channel 1's low frequencies.*
Highs1	*Controls the level of channel 1's high frequencies.*
Lows2	*Controls the level of channel 2's low frequencies.*
Highs2	*Controls the level of channel 2's high frequencies.*
Low1Pan	*Controls where in the output stereo field channel 1's low frequencies end up.*
Hi1Pan	*Controls where in the output stereo field channel 1's high frequencies end up.*
Low2Pan	*Controls where in the output stereo field channel 2's low frequencies end up.*
Hi2Pan	*Controls where in the output stereo field channel 2's high frequencies end up.*

dual 4band para

(131   Dual 4B Parametric)

This stereo four-band equalizer has 'master' and individual channel parameters. In addition, band #1 is selectable between lowpass and bandpass, while band #4 can be either bandpass or highpass.
Stereo in, stereo out

Masters		
Band#1		
Level1	*Controls the boost or cut of band 1.*	
Freq1	*Controls the center frequency or cut-off frequency of band 1.*	
Q1	*Controls the bandwidth or resonance of band 1.*	
Type1	*Controls whether band one behaves as a bandpass filter or as a lowpass filter (other bands are either dedicated bandpass filters or allow for highpass filtering as well).*	
Band#2		
Parameters similar to those described above, dedicated to band 2.		
Band#*3*		
Parameters similar to those described above, dedicated to band 3.		
Band#4		
Parameters similar to those described above, dedicated to band 4.		

EQ#1
Parameters similar to those described above, dedicated to EQ 1; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

EQ#2
Parameters similar to those described above, dedicated to EQ 2; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

dual 8 band EQ

(132   Dual8Band Eq)

This stereo eight-band equalizer has 'master' and individual channel parameters as well as a single bandwidth parameter.

Stereo in, stereo out

See "dual 4band para" above for diagram and parameters.

dual filters

(133   Dual Filters)

This algorithm offers 'master' and individual channel parameters. The filter type may be selected from lowpass, hipass, bandpass or notch, while the slope may be either 24 or 48 dB per octave.

Stereo in, stereo out

Master	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the input to the loop with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
Gain	*Controls the output gain of the filters.*
Freq	*Controls the center frequency or cutoff frequency.*
Q	*Controls the bandwidth or resonance.*
Type	*Selects the type of filter.*
Order	*Selects the "steepness" of the filter.*

Filter#1
Parameters similar to those described above, dedicated to Filter 1; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

Filter#2
Parameters similar to those described above, dedicated to Filter 2; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

dual modfilters

(134   Dual Modfilters)

This modulated filter algorithm offers 'master' and individual channel parameters, with a 'mode' selector for modulation (one of *LFO*, envelope or pedal), as well as 'freq' and 'q' modulation. The envelope follower may be triggered from either input.
Stereo in, stereo out

Master	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the input to the loop with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
Mode	*Selects whether the LFO, the envelope generator, or an external pedal will modulate the filters.*
Rate	*Controls the LFO modulation rate if T_Rate1 is set to "off" or reflects the physical LFO modulation rate as selected by T_Rate and the system tempo.*
T_Rate	*Controls the LFO modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic LFO modulation generation "off".*
Type	*Selects the type of filter.*
FAttack	*Controls how quickly the filter frequency will track an increase in the modulation source level.*
FDecay	*Controls how quickly the filter frequency will track a decrease in the modulation source level.*
QAttack	*Controls how quickly the filter Q will track an increase in the modulation source level.*
QDecay	*Controls how quickly the filter Q will track a decrease in the modulation source level.*
Freq	*Controls the "baseline" center frequency or cutoff frequency.*
Fmod	*Controls how far from the "baseline" the cutoff frequency will deviate in response to the modulation signal.*
Q	*Controls the "baseline" bandwidth or resonance.*
Qmod	*Controls how far from the "baseline" the Q will deviate in response to the modulation signal.*

Env	
Env	*Determines which input or inputs the envelope follower will track.*
Sens	*Controls how sensitive the envelope follower is; alter this parameter to change the "depth" of the FM effect.*

Retrig   *Whenever the LFO Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.*

Angle   *Given most configurations of the Eclipse, LFO modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the LFO begins.*

Filter#1

Most parameters similar to those described above, dedicated to Filter 1; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")
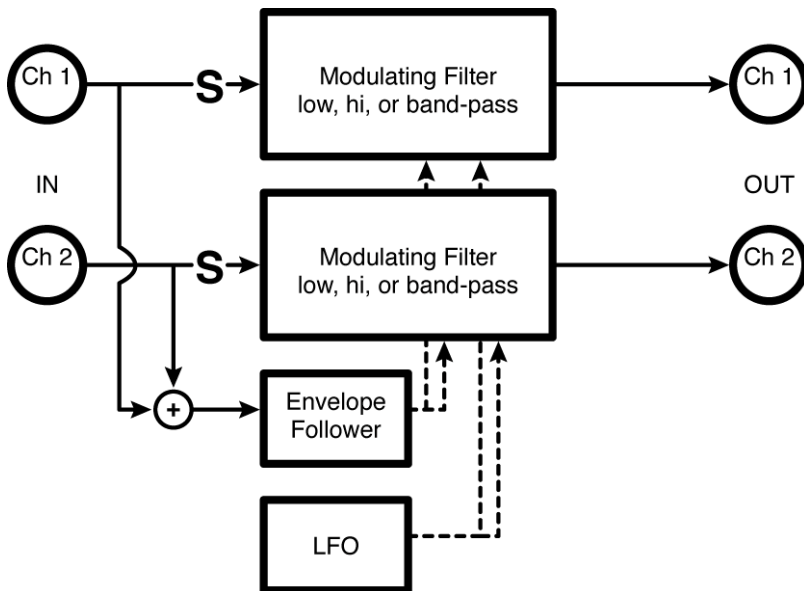
Man_Ped1   *Allows users without an external pedal to modulate the effect from the front panel.*

Filter#2

Most parameters similar to those described above, dedicated to Filter 2; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

Man_Ped1   *Allows users without an external pedal to modulate the effect from the front panel.*

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stereo phaser

(135   St Phaser)

This simple stereo phase shifter allows the number of poles to be selected from 3 to 24, allowing great control over the strength of the effect. A 'mode' selector is provided for choice of modulation input between *LFO* and pedal.

Stereo in, stereo out

Phaser

Send   *Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*

Mix   *Controls the ratio between the dry and wet (phased) signal. Set to "50%" if the dry signal won't be mixed in later (e.g. in a guitar rig).*

Mode   *Selects whether the LFO or an external pedal will modulate the filters.*

Depth   *Controls the depth of the phase effect.*

Fback   *Controls how much of each phaser's output is reapplied to its input for exaggerated phasing.*

#Poles   *Determines how many "poles" the phaser's filters will use. More poles: more pronounced effect.*

Sweep   *Determines if the modulation signal to each phaser is in phase ("sync") or out of phase ("oppose").*

Man_Ped   *Allows users without an external pedal to modulate the effect from the front panel.*

LFO

Rate   *Controls the modulation rate if T_Rate1 is set to "off" or reflects the physical modulation rate as selected by T_Rate and the system tempo.*

T_Rate *Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".*

Shape *Determines the shape of the modulating signal.*

Duty *Controls the "duty cycle" of the modulating wave for any waveform other than "sine". For example, if you have selected "triangle" under Shape and set Duty to "0%", the triangle will morph to a sawtooth (very fast rise, slow fall). With Duty at "100%", the triangle will morph to a ramp (slow rise, very fast fall).*

Retrig *Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.*

Angle *Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.*

stereoizing phaser

(136   StereoizingPhaser)

Two phase shifters tapped off the 9th pole (left) and the 12th pole (right) create a credible stereo field.
Summed in, stereo out

Phaser

Send *Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*

Mix *Controls the ratio between the dry and wet (phased) signal. Set to "50%" if the dry signal won't be mixed in later (e.g. in a guitar rig).*

Mode *Selects whether the LFO or an external pedal will modulate the filters.*

Depth *Controls the depth of the phase effect.*

Fback *Controls how much of each phaser's output is reapplied to its input for exaggerated phasing.*

Man_Ped *Allows users without an external pedal to modulate the effect from the front panel.*

LFO

Rate *Controls the modulation rate if T_Rate1 is set to "off" or reflects the physical modulation rate as selected by T_Rate and the system tempo.*
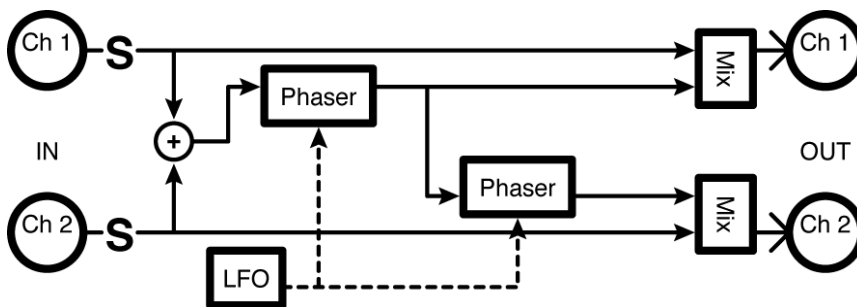
T_Rate *Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".*

Shape *Determines the shape of the modulating signal.*

Duty *Controls the "duty cycle" of the modulating wave for any waveform other than "sine". For example, if you have selected "triangle" under Shape and set Duty to "0%", the triangle will morph to a sawtooth (very fast rise, slow fall). With Duty at "100%", the triangle will morph to a ramp (slow rise, very fast fall).*

Retrig   *Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.*

Angle   *Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.*

dual sample/hold

(137 DualSamp/Hold)

This unusual effect features dual low pass filters driven by a rhythmic control. It features 'master' and individual channel parameters. In addition there are controls for 'min/max' frequency and bandwidth as well as 'resonance'. The 'flux' parameter smoothes out changes.
Stereo in, stereo out

```
Ch 1 ──────►┌──────────────────┐
            │ Modulating Low-pass│───►( + )───► Ch 1
            │      Filter        │
IN          └──────────────────┘              OUT
Ch 2 ──────►┌──────────────────┐
            │ Modulating Low-pass│───►( + )───► Ch 2
            │      Filter        │
            └──────────────────┘

┌─────┐      ┌──────────┐
│ LFO │─ ─ ─►│  Sample  │
└─────┘      │  & Hold  │
             └──────────┘
```

Master

Send   *Adjusts the level at 'S' in the diagram. If you want to modulate the input to the loop with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*

Mix   *Controls the ratio between the dry and wet (S/H) signal.*

***F* Min**   *Sets the minimum frequency for the Sample and Hold.*

***F* Max**   *Sets the maximum frequency for the Sample and Hold.*

Reson   *Controls the filter resonance.*

Flux   *Controls how quickly the filter moves between "notes".*

Rate   *Controls the modulation rate if T_Rate is set to "off" or reflects the physical modulation rate as selected by T_Rate and the system tempo.*

T_Rate   *Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".*

Retrig   *Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.*
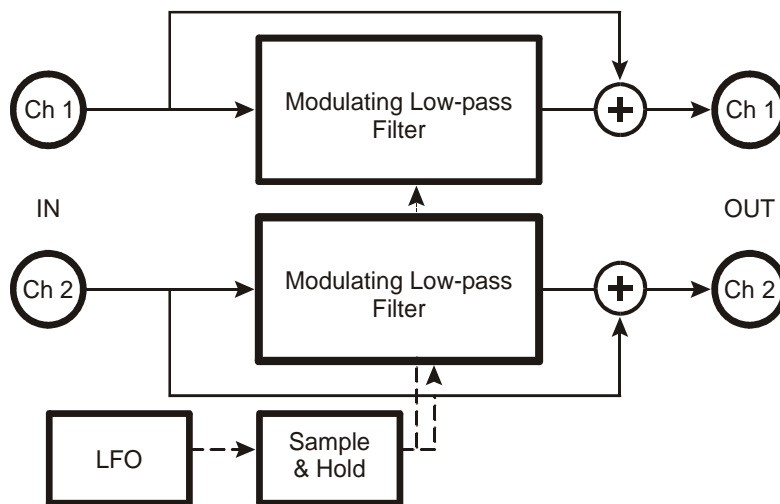
Angle   *Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.*

Eclipse Algorithms Manual

S/H#1

Parameters similar to those described above, dedicated to Sample and Hold 1; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")
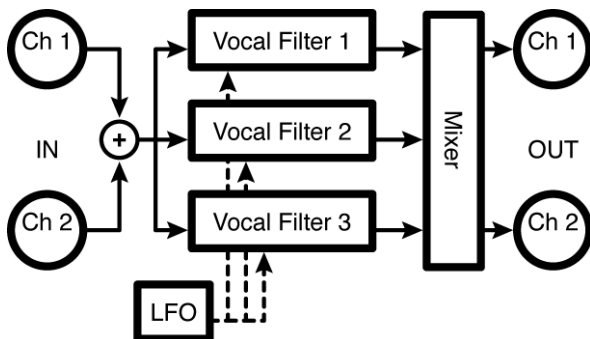
S/H#2

Parameters similar to those described above, dedicated to Sample and Hold 2; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

vocal wa

(138   VocalizedWa)

This notorious Eventide effect contains three filter channels used as 'vowel' or 'formant' selectors, with individual level and pan controls. In addition, a 'mode' selector allows modulation from either an *LFO* or a pedal.

Summed in, stereo out

Vowels	
Mode	Selects whether the LFO or an external pedal will modulate the filters.
Vowel1	Selects one of the filter "vowels".
Vowel2	Selects the second filter "vowel".
Q	Controls the bandwidth of the filters.
QMod	Controls how much the Q changes with modulation.
Man_Ped	Allows users without an external pedal to modulate the effect from the front panel.

LFO	
Rate	Controls the modulation rate if T_Rate1 is set to "off" or reflects the physical modulation rate as selected by T_Rate and the system tempo.
T_Rate	Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".
Shape	Determines the shape of the modulating signal.
Duty	Controls the "duty cycle" of the modulating wave for any waveform other than "sine". For example, if you have selected "triangle" under Shape and set Duty to "0%", the triangle will morph to a sawtooth (very fast rise, slow fall). With Duty at "100%", the triangle will morph to a ramp (slow rise, very fast fall).
Retrig	Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.

© 2001 Eventide, Inc.   Page 34 of 79   Manual Version 1.1

Angle   *Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.*

Mixer^(x = 1, 2, 3)

Level x   *Controls the level of filter x.*

Pan x   *Positions the image of filter x in the stereo field.*

---

10 band vocoder

(139 Vocoder10)

A vocoder allows the characteristics of one signal to be superimposed on another; for example a guitar can be made to 'sing,' or a choir-like effect can be produced by using a noise source as the instrument.

The left input (channel #1) or and internal noise source feeds carrier (instrument), while right input (channel #2) feeds the modulator (voice). The output is the sum of ten frequency bands from 20Hz to 20kHz, with 'level' and 'pan' for each band.
Dual in, stereo out

Cal

Carrier   *Determines if the spectrum supplied by input 2 (voice) will be applied to "noise" or to the signal supplied at input 1 ("left in").*

L Gain   *Controls the level of the left input (instrument).*

R Gain   *Controls the level of the right input (voice).*

Q/Res   *Controls the bandwidth of each filter. Lower values deliver a louder, albeit less distinct effect.*

Attack   *Controls how quickly the envelope follower on each filter will track an increase in the right input (voice).*

Decay   *Controls how quickly the envelope follower on each filter will track a decrease in the right input (voice).*

Freqs^(x = 1, 2, ..., 10)

Freq x   *Controls the center frequency for band filter x. Avoid centering on frequencies outside the range of the right input (voice) - it's just a waste of a filter!*

Mixer

Levels^(x = 1, 2, ..., 10)

M_Level   *Scales all the individual band filter levels, effectively serving as a master volume.*

Levelx   *Controls the level of band filter* x.

Pans^(x = 1, 2, ..., 10)

M_Pan   *Scales all the individual band filter pans, effectively controlling the stereo spread.*

Panx   *Controls the stereo placement of band filter* x.

plex

A plex *is the feedback* network *of a reverb.*

large delay 8 plex

(141   LrgDelay8Plex)

This plex has eight two second modulating delay lines with 'level' and 'pan' parameters for each independent output.
Summed in, stereo out

Masters	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
Level	*Controls the output level.*
Decay	*Controls the delay feedback, effectively the reverb time.*
Size	*Scales the individual delay times, effectively altering the "room size".*
HiCut	*Controls the output level of a high shelving filter.*
LowCut	*Controls the output level of a low shelving filter.*
HiFreq	*Controls the cutoff frequency of the high shelving filter.*
LowFreq	*Controls the cutoff frequency of the low shelving filter.*
Depth	*Controls the depth of delay modulation for a chorus effect.*
Rate	*Controls the modulation rate if T_Rate1 is set to "off" or reflects the physical modulation rate as selected by T_Rate and the system tempo.*
T_Rate	*Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".*
Image	*Scales all the individual delay pans, effectively controlling the stereo spread.*
Retrig	*Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.*

Angle   Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.

Plex

Delays**^(x = *1, 2, ..., 8*)**

Delayx   *Sets the delay time for delay* x.

Mixer**^(x = *1, 2, ..., 8*)**

Levelx   *Controls the level of delay* x.

Panx   *Controls the stereo placement of delay* x.

---

delay 8 plex

(140   Delay8Plex)

This one has eight 660 mS modulating delay lines with 'level' and 'pan' parameters for each independent output.
Summed in, stereo out

See "large delay 8 plex" above for diagram and parameters.

---

detune 4 plex

(142   Detune4Plex)

This one has four 660 mS *detuner*s in addition to 'level' and 'pan' parameters for each independent output.
Summed in, stereo out

Masters

Send   *Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*

Level   *Controls the output level.*

Decay   *Controls the delay feedback, effectively the reverb time.*

Size   *Scales the individual delay times, effectively altering the "room size".*

HiCut   *Controls the output level of a high shelving filter.*

LowCut   *Controls the output level of a low shelving filter.*

HiFreq   *Controls the cutoff frequency of the high shelving filter.*

LowFreq	*Controls the cutoff frequency of the low shelving filter.*
Detune	*Scales all the individual detune amounts, effectively serving as a master detune.*
Length	*Controls the delay time of the detune pitchshifters.*
Image	*Scales all the individual delay pans, effectively controlling the stereo spread.*

Detune^(x = 1, 2, 3, 4)

Detunex	*Controls the pitchshift of detuner x.*
Delayx	*Controls the delay time of pitchshift x.*

Mixer^(x = 1, 2, 3, 4)

Levelx	*Controls the level of detuner x.*
Panx	*Controls the stereo placement of detuner x.*

reverse 4 plex

(143   Reverse4Plex)

This one adds four two-second reverse shifters as well as 'level' and 'pan' parameters for each independent output.
Summed in, stereo out

See "detune 4 plex" above for diagram.

Masters

Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
Level	*Controls the output level.*
Decay	*Controls the reverse feedback.*
Size	*Scales the individual reverse times, effectively altering the "room size".*
Pitch	*Scales all the individual reverse shifters.*
HiCut	*Controls the output level of a high shelving filter.*
LowCut	*Controls the output level of a low shelving filter.*
HiFreq	*Controls the cutoff frequency of the high shelving filter.*
LowFreq	*Controls the cutoff frequency of the low shelving filter.*
Image	*Scales all the individual delay pans, effectively controlling the stereo spread.*

Plex

Pitch^(x = 1, 2, 3, 4)

Pitchx	*Controls the pitch of reverse shifter x.*

Delays^(x = 1, 2, 3, 4)

Delayx	*Controls the delay time of reverse shifter x.*

Mixer^(x = 1, 2, 3, 4)

Levelx	*Controls the level of reverse shifter x.*
Panx	*Controls the stereo placement of reverse shifter x.*

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preamps

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fuzzADSRpre

(144   FuzzADSRPreamp)

This instrument preamp has compression, by-passable fuzz, EQ, remote 'volume pedal' and gate, as well as an ADSR controlled amp and an ASDR controlled filter for synth emulation.
Summed in, summed out

PreAmp	
Comp	
Thresh	*Sets the input threshold above which compression takes place and below which the signal is left alone.*
Gain	*Adjusts the output level to improve gain structure.*
Attack	*Controls how quickly the compressor "clamps down" on a signal after it exceeds the threshold.*
Decay	*Controls how quickly the compressor "lets go" of a signal once it falls below the threshold.*
Ratio	*Controls the input to output ratio for signals above the threshold. For example, "6:1" produces 1dB of output for every 6dB of input.*
Fuzz	
Fuzz	*Selects distortion "in" or "out".*
Drive	*Controls how much input (drive) is sent to the distortion. The greater the input, the greater the distortion.*
Level	*Controls the output level of the distortion.*
Flux	*Controls the distortion "slew rate" to change its timbre, or how "in your face" it sounds.*
HiCut	*Controls the cutoff frequency of a gentle lowpass filter.*
EQ	
Trim	*Controls the input level to the EQ.*
Low	*Controls the low EQ boost/cut.*
Mid	*Controls the mid EQ boost/cut.*
High	*Controls the high EQ boost/cut.*
Low	*Controls the low EQ center frequency.*
Mid	*Controls the mid EQ center frequency.*
High	*Controls the high EQ center frequency.*

Manual_P   *Allows users without an external pedal to modulate the effect from the front panel.*

Eclipse Algorithms Manual

Gate	
G Thresh	*Sets the input threshold that the signal must exceed to open the gate.*
G Attack	*Controls how quickly the gate opens up once the signal exceeds the threshold.*
G Decay	*Controls how quickly the gate closes once the signal falls below the threshold.*

A ADSR Controls the ADSR which modulates the Amp	
Sens	*Controls the sensitivity of the A and F ADSR triggers.*
Retrig	*Determines the crossfade time between the old modulation level and the new modulation level when the A ADSR is retriggered.*
A DLevel	*Controls the level of the initial A ADSR peak as a percentage of the total possible modulation output.*
A SLevel	*Controls the level of the sustain portion of the A ADSR as a percentage of the total possible modulation output.*
A Attack	*Controls the attack time.*
A Decay	*Controls the decay time.*
A Sustain	*Controls the sustain time.*
A Releas	*Controls the release time.*

Peak	
Attack	*Smoothes the rise time of the signal that triggers the A and F ADSRs.*
Decay	*Smoothes the fall time of the signal that triggers the A and F ADSRs.*

Filter	

Filter	
Freq	*Controls the "baseline" center frequency or cutoff frequency.*
Fmod	*Controls how far from the "baseline" the cutoff frequency will deviate in response to the F ADSR modulation signal.*
Q	*Controls the "baseline" bandwidth or resonance.*
Qmod	*Controls how far from the "baseline" the Q will deviate in response to the F ADSR modulation signal.*
Type	*Selects the type of filter.*

F ADSR – controls the Filter ADSR	
Sens	*Controls the sensitivity of the A and F ADSR triggers – this is the same control as above, duplicated for convenience.*
Retrig	*Determines the crossfade time between the old modulation level and the new modulation level when the F ADSR is retriggered.*
F DLevel	*Controls the level of the initial F ADSR peak as a percentage of the total possible modulation output.*
F SLevel	*Controls the level of the sustain portion of the F ADSR as a percentage of the total possible modulation output.*
F Attack	*Controls the attack time.*
F Decay	*Controls the decay time.*
F Sustain	*Controls the sustain time.*
F Releas	*Controls the release time.*

Peak - same controls as above, duplicated for convenience	
Attack	*Smoothes the rise time of the signal that triggers the A and F ADSRs.*
Decay	*Smoothes the fall time of the signal that triggers the A and F ADSRs.*

Eclipse Algorithms Manual

bass pre

(145   BassPreamp)

A simple preamp configuration with compression, three-band EQ, class A distortion, and a 'thickener'
parameter.
Summed in, summed out

Comp	
Thresh	*Sets the input threshold above which compression of the signal takes place and below which the signal is left alone.*
Ratio	*Controls the input to output ratio for signals above the threshold. For example, "6:1" produces 1dB of output for every 6dB of input.*
Attack	*Controls how quickly the compressor "clamps down" on a signal after it exceeds the threshold.*
Decay	*Controls how quickly the compressor "lets go" of a signal once it falls below the threshold.*

Tone	
Trim	*Controls the input level to the EQ.*
Low	*Controls the low EQ boost/cut.*
Mid	*Controls the mid EQ boost/cut.*
High	*Controls the high EQ boost/cut.*
Thicken	*Controls the amount of "thickener". Slather it on...*
Distort	*Controls how much input (drive) is sent to the distortion. The greater the input, the greater the distortion.*

overdrive preamp

(146   OverdrivePreamp)

Instrument preamp with compression, bypassable fuzz, EQ, remote 'volume pedal' and gate. This flavor is an overdrive effect and is much more interactive with the source signal.
Summed in, summed out

PreAmp

OD

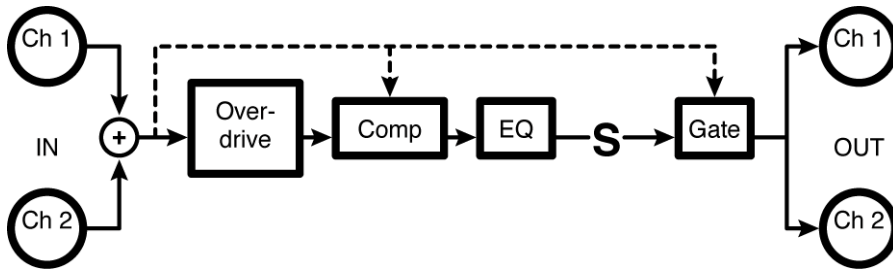|	
Fuzz	*Selects distortion "in" or "out".*
Drive	*Controls how much input (drive) is sent to the distortion. The greater the input, the greater the distortion.*
Level	*Controls the output level of the distortion.*
Flux	*Controls the distortion "slew rate" to change its timbre, or how "in your face" it sounds.*
HiCut	*Controls the cutoff frequency of a gentle lowpass filter.*

Comp

Thresh	*Sets the input threshold above which compression takes place and below which the signal is left alone.*
Gain	*Adjusts the output level to improve gain structure.*
Attack	*Controls how quickly the compressor "clamps down" on a signal after it exceeds the threshold.*
Decay	*Controls how quickly the compressor "lets go" of a signal once it falls below the threshold.*
Ratio	*Controls the input to output ratio for signals above the threshold. For example, "6:1" produces 1dB of output for every 6dB of input.*

EQ

Trim	*Controls the input level to the EQ.*
Mid1	*Controls the mid1 EQ center frequency.*
Mid1	*Controls the mid1 EQ boost/cut.*
Mid2	*Controls the mid2 EQ center frequency.*
Mid2	*Controls the mid2 EQ boost/cut.*

Gate

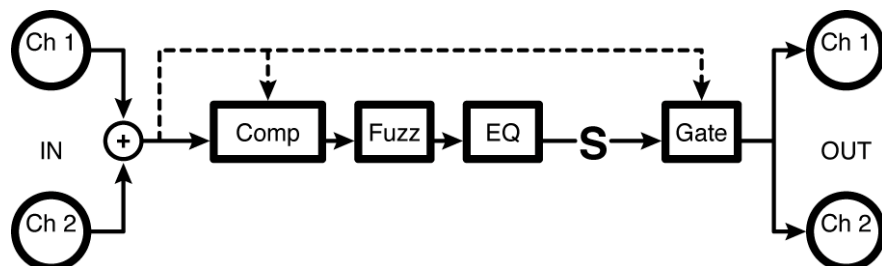|	
G Thresh	*Sets the input threshold that the signal must exceed to open the gate.*
G Attack	*Controls how quickly the gate opens up once the signal exceeds the threshold.*
G Decay	*Controls how quickly the gate closes once the signal falls below the threshold.*

Manual_P   *Allows users without an external pedal to modulate the effect from the front panel.*

Eclipse Algorithms Manual

fuzz preamp

(147   FuzzPreamp)

This instrument preamp has compression, bypass-able fuzz, EQ, remote 'volume pedal' and gate.
Summed in, summed out

PreAmp

Comp

Thresh	*Sets the input threshold above which compression takes place and below which the signal is left alone.*
Gain	*Adjusts the output level to improve gain structure.*
Attack	*Controls how quickly the compressor "clamps down" on a signal after it exceeds the threshold.*
Decay	*Controls how quickly the compressor "lets go" of a signal once it falls below the threshold.*
Ratio	*Controls the input to output ratio for signals above the threshold. For example, "6:1" produces 1dB of output for every 6dB of input.*

Fuzz

Fuzz	*Selects distortion "in" or "out".*
Freq	*Controls the frequency of a pre-drive band EQ.*
Filter	*Controls the mix of the pre-drive filter.*
Q	*Controls the bandwidth or resonance of the pre-drive filter.*
Drive	*Controls how much input (drive) is sent to the distortion. The greater the input, the greater the distortion.*
Level	*Controls the output level of the distortion.*
Flux	*Controls the distortion "slew rate" to change its timbre, or how "in your face" it sounds.*
HiCut	*Controls the cutoff frequency of a gentle lowpass filter.*

EQ

Trim	*Controls the input level to the EQ.*
Low	*Controls the low EQ boost/cut.*
Mid	*Controls the mid EQ boost/cut.*
High	*Controls the high EQ boost/cut.*
Low	*Controls the low EQ center frequency.*
Mid	*Controls the mid EQ center frequency.*
High	*Controls the high EQ center frequency.*

Gate	
G Thresh	*Sets the input threshold that the signal must exceed to open the gate.*
G Attack	*Controls how quickly the gate opens up once the signal exceeds the threshold.*
G Decay	*Controls how quickly the gate closes once the signal falls below the threshold.*
Manual_P	*Allows users without an external pedal to modulate the effect from the front panel.*

polydriver

(148   PolyDriver)

A compressor feeds six bandpassed class A drivers mixed to stereo leading to a remote 'volume pedal' and gate.
Summed in, stereo out

PreAmp	

Comp	
Thresh	*Sets the input threshold above which compression takes place and below which the signal is left alone.*
Gain	*Adjusts the output level to improve gain structure.*
Attack	*Controls how quickly the compressor "clamps down" on a signal after it exceeds the threshold.*
Decay	*Controls how quickly the compressor "lets go" of a signal once it falls below the threshold.*
Ratio	*Controls the input to output ratio for signals above the threshold. For example, "6:1" produces 1dB of output for every 6dB of input.*

Drives	
Trim	*Controls the input level to the distortion.*
Drive 1..6	*Controls the drive level of drive* 1 to 6.

Filters	
Freq 1..6	*Controls the center frequency for band filter* 1 thru 6.
Q	*Controls the bandwidth for all six band filters.*
HiCut	*Controls the cutoff frequency of a gentle lowpass filter.*

Pans	
Pan 1..6	*Controls the stereo position for band* 1 thru 6.

fuzzpre wa/wammy

(149   FuzzWaWammyPre)

This is an instrument preamp with compression, bypass-able fuzz, EQ, remote 'volume pedal' and gate. This flavor has 'wah-wah' and 'wammy' effects. You can choose which effect is controlled by the 'pedal.'
Summed in, summed out

PreAmp

Comp

Thresh	*Sets the input threshold above which compression takes place and below which the signal is left alone.*
Gain	*Adjusts the output level to improve gain structure.*
Attack	*Controls how quickly the compressor "clamps down" on a signal after it exceeds the threshold.*
Decay	*Controls how quickly the compressor "lets go" of a signal once it falls below the threshold.*
Ratio	*Controls the input to output ratio for signals above the threshold. For example, "6:1" produces 1dB of output for every 6dB of input.*

Fuzz

Fuzz	*Selects distortion "in" or "out".*
Drive	*Controls how much input (drive) is sent to the distortion. More input, more distortion.*
Level	*Controls the output level of the distortion.*
Flux	*Controls the distortion "slew rate" to change its timbre, or how "in your face" it sounds.*
HiCut	*Controls the cutoff frequency of a gentle lowpass filter.*

EQ

Trim	*Controls the input level to the EQ.*
Low	*Controls the low EQ boost/cut.*
Mid	*Controls the mid EQ boost/cut.*
High	*Controls the high EQ boost/cut.*
Low	*Controls the low EQ center frequency.*
Mid	*Controls the mid EQ center frequency.*
High	*Controls the high EQ center frequency.*

Gate

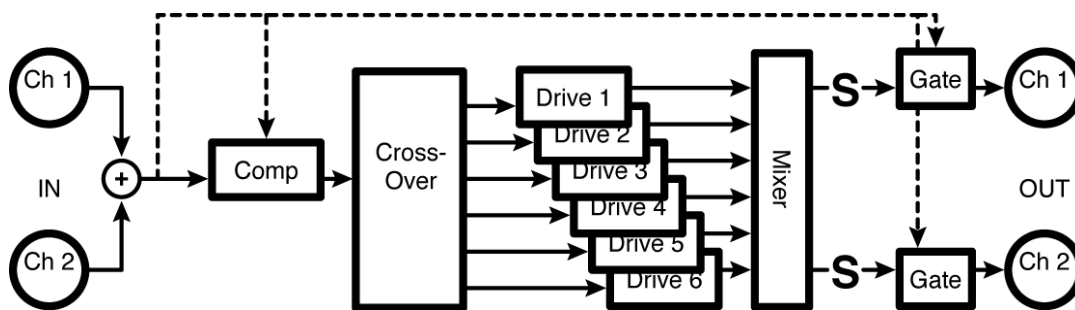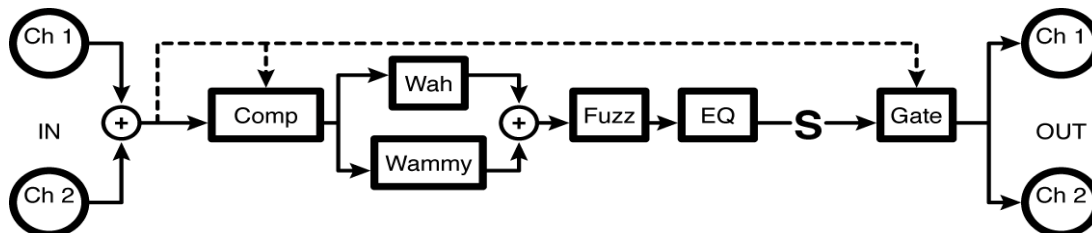|	
G Thresh	*Sets the input threshold that the signal must exceed to open the gate.*
G Attack	*Controls how quickly the gate opens up once the signal exceeds the threshold.*
G Decay	*Controls how quickly the gate closes once the signal falls below the threshold.*

Pedals

Effect   *Select which effect the external foot pedal will control: "volume", "wammy" (pitch shifter), or "wawa".*
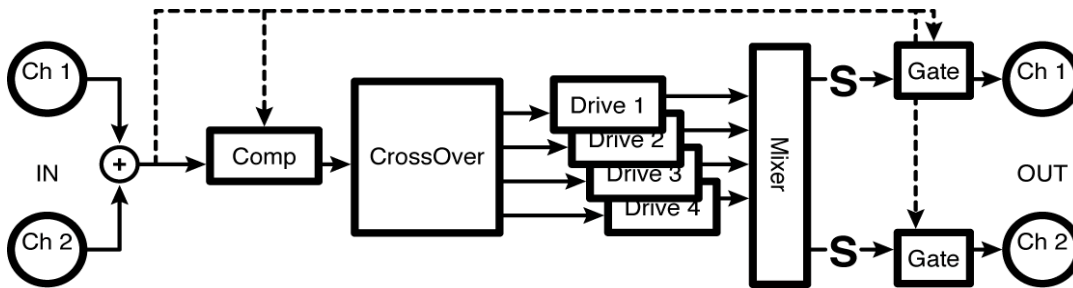
Heel	*Determines the "wammy" pitch when the pedal is at the heel position.*
Toe	*Determines the "wammy" pitch when the pedal is at the toe position.*
Delay	*Controls the "wammy" pitchshifter delay time.*
Freq	*Controls the "baseline" center frequency or cutoff frequency for the "wawa".*
Fmod	*Controls how far from the "baseline" the cutoff frequency will deviate in response to the modulation signal for the "wawa".*
Q	*Controls the "baseline" bandwidth or resonance for the "wawa".*
Qmod	*Controls how far from the "baseline" the Q will deviate in response to the modulation signal for the "wawa".*
LowNote	*Determines the lowest note the "wammy" pitchshifter will correctly shift. Set as high as possible for best results.*
XFade	*Controls the crossfade between old "wammy" pitchshift values and new pitchshift values.*
Manual_P	*Allows users without an external pedal to modulate the effect from the front panel.*

ez polyfuzz

(150   EZPolyFuzz)

A compressor feeds a four-band *polyfuzz* (each band feeds individual distortion elements). This one has simplified parameters.
Summed in, stereo out

PreAmp

Comp

Thresh	*Sets the input threshold above which compression takes place and below which the signal is left alone.*
Gain	*Adjusts the output level to improve gain structure.*
Attack	*Controls how quickly the compressor "clamps down" on a signal after it exceeds the threshold.*
Decay	*Controls how quickly the compressor "lets go" of a signal once it falls below the threshold.*
Ratio	*Controls the input to output ratio for signals above the threshold. For example, "6:1" produces 1dB of output for every 6dB of input.*

PolyFuzz

Tone	*Controls the overall tone of the distortion (higher values are brighter).*
Q	*Controls the bandwidth of all the band filters.*
Drive	*Controls how much input (drive) is sent to the distortion. More input, more the distortion.*
Flux	*Controls the distortion "slew rate" to change its timbre, or how "in your face" it sounds.*
Level	*Controls the output level of the distortion.*
Image	*Controls the stereo spread of the effect.*
HiCut	*Controls the cutoff frequency of a gentle lowpass filter.*

<table>
<tr><td colspan="2"><i>Gate</i></td></tr>
<tr><td>G Thresh</td><td><i>Sets the input threshold that the signal must exceed to open the gate.</i></td></tr>
<tr><td>G Attack</td><td><i>Controls how quickly the gate opens up once the signal exceeds the threshold.</i></td></tr>
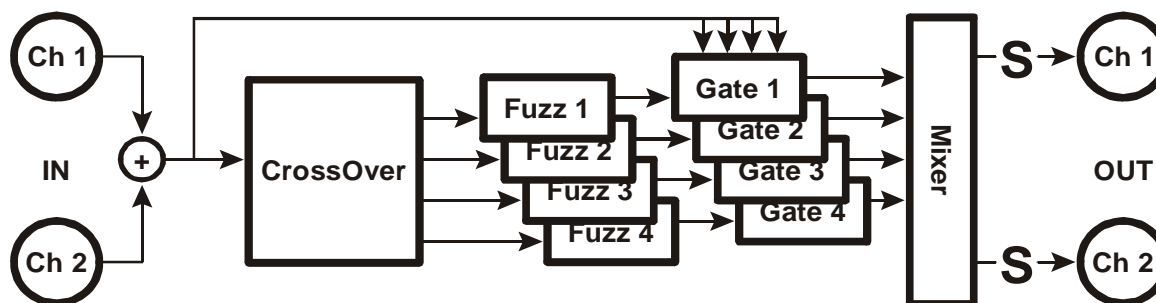<tr><td>G Decay</td><td><i>Controls how quickly the gate closes once the signal falls below the threshold.</i></td></tr>
</table>

Manual_P   *Allows users without an external pedal to modulate the effect from the front panel.*

polyfuzz

(151   PolyFuzz)

Here the compressor feeds a four band gated *polyfuzz* (each band feeds individual distortion elements). This one has individual parameters.
Summed in, stereo out

<table>
<tr><td colspan="2"><i>PolyFuzz</i></td></tr>
</table>

<table>
<tr><td colspan="2"><i>Filters^(x = 1, 2, 3, 4)</i></td></tr>
<tr><td>Freq x</td><td><i>Controls the center frequency for band filter</i> x.</td></tr>
<tr><td>Q x</td><td><i>Controls the bandwidth for band filter</i> x.</td></tr>
</table>

<table>
<tr><td colspan="2"><i>Fuzz^(x = 1, 2, 3, 4)</i></td></tr>
<tr><td>Drive x</td><td><i>Controls how much input (drive) from band filter</i> x <i>is sent to the distortion. The greater the input, the greater the distortion.</i></td></tr>
<tr><td>Level x</td><td><i>Controls the output level of drive</i> x.</td></tr>
</table>

<table>
<tr><td colspan="2"><i>HiCuts^(x = 1, 2, 3, 4)</i></td></tr>
<tr><td>Flux x</td><td><i>Controls the distortion "slew rate" of drive</i> x <i>to change its timbre, or how "in your face" it sounds.</i></td></tr>
<tr><td>HiCut x</td><td><i>Controls the cutoff frequency of a gentle lowpass filter after drive</i> x.</td></tr>
</table>

<table>
<tr><td colspan="2"><i>Image^(x = 1, 2, 3, 4)</i></td></tr>
<tr><td>Pan x</td><td><i>Controls where in the stereo field drive</i> x <i>will appear (negative values scoot it left and positive values scoot it right).</i></td></tr>
</table>

<table>
<tr><td colspan="2"><i>Gates^(x = 1, 2, 3, 4)</i></td></tr>
<tr><td>Thresh</td><td><i>Sets the input threshold that the signal must exceed to open the gate.</i></td></tr>
<tr><td>Attack x</td><td><i>Controls how quickly the gate after drive</i> x <i>opens up once the signal exceeds the threshold.</i></td></tr>
<tr><td>Decay x</td><td><i>Controls how quickly the gate after drive</i> x <i>closes once the signal falls below the threshold.</i></td></tr>
</table>

Manual_P   *Allows users without an external pedal to modulate the effect from the front panel.*

shifters

A diatonic pitch shifter shifts the musical pitch of an audio signal while maintaining the proper harmonic relationship to a diatonic scale. To accomplish this, the user specifies the desired key, scale and musical interval. The pitch shifter takes care of finding out what note is being played and automatically adjusts the amount of pitch shift so that the resultant note is in-key.
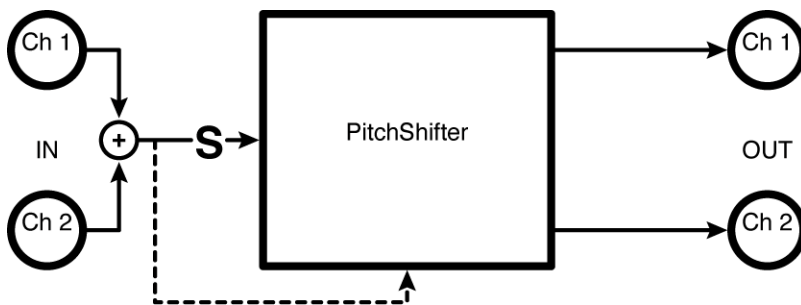
In general, the operation of any shifter can be improved by optimizing the setting of the 'low note' parameter, which defines the lowest note that the shifter recognizes. This should be set low for polyphonic material such as guitar, and higher for vocals.

The 'xfade' parameter can usually be left alone – greater values may increase smoothness at the cost of blurring transients, lower values may add roughness.

auto pitch correct

(158   AutoCorrect)

This algorithm is essentially a diatonic shifter with independent delay parameters. Set as 'chromatic.'
Summed in, stereo out

Meter	
Error	*Monitors how far off the Eclipse believes the input is from "correct".*
Fixed	*Monitors how far the Eclipse is bending (or fixing) the input to correct the error.*

Shift	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
Quant	*Determines the level of "quantization", or how close the Eclipse will bring an "off" note to "perfection".*
Tune	*Controls the detector's calibration, allowing for deviation from A-440 tuning.*
Key	*Selects the key that the corrector references (if a scale is in fact selected under Scale).*
Scale	*Selects the scale that the corrector references.*
Glide	*Controls how quickly the shifter slides between shift values.*
Delay L	*Controls the left channel shifter delay time.*
Delay R	*Controls the right channel shifter delay time.*
Level L	*Controls the left channel output level.*
Level R	*Controls the right channel output level.*

Cal

LowNote   *Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.*

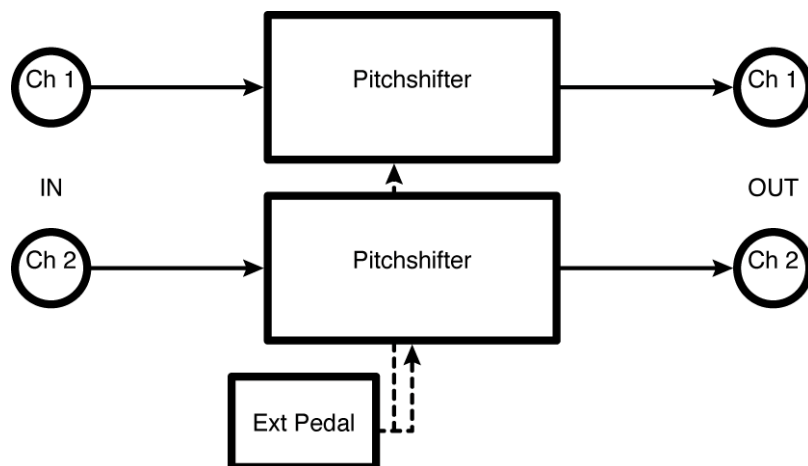**Source**   *Optimizes the corrector for either "vocal" or "instrument" input.*

XFade   *Controls the crossfade between old pitchshift values and new pitchshift values.*

---

dual wammy

(159   DualWammy)

This externally controlled shifter has 'master' and individual channel parameters. Pitch and modulation are set up as 'heel/toe' parameters.

Stereo in, stereo out

Master

Bypass   *Selects "fx in" or "fx out".*

Man_Wam   *Allows users without an external pedal to modulate the effect from the front panel.*

Heel   *Determines the "wammy" pitch when the pedal is at the heel position.*

Toe   *Determines the "wammy" pitch when the pedal is at the toe position.*

Delay   *Controls the "wammy" pitchshifter delay time.*

LowNote   *Determines the lowest note the "wammy" pitchshifter will correctly shift. Set as high as possible for best results.*

XFade   *Controls the crossfade between old "wammy" pitchshift values and new pitchshift values.*

Wammy#1

Parameters similar to those described above, dedicated to "wammy" pitchshifter 1; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

Wammy#2

Parameters similar to those described above, dedicated to "wammy" pitchshifter 2; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")
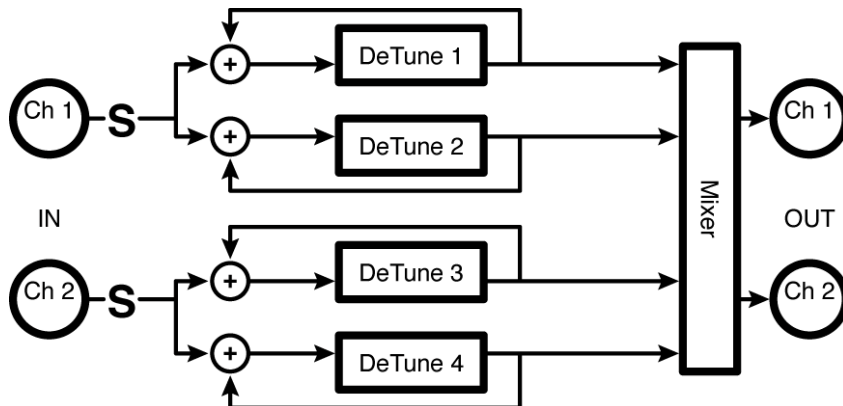
4 detune*r*s

(160   4Detuners)

There are four 660 mS *detuner*s with independent feedback paths. 'Tightness' is a master length/splice parameter.
Stereo in, stereo out

Masters	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
m_Level	Scales all the individual detune levels, effectively serving as a master volume.
m_Pitch	Scales all the individual detune amounts, effectively serving as a master pitch.
m_Delay	Scales all the individual detune delay times, effectively serving as a master delay (e.g. if the four delays are set to 200ms, 150ms, 100ms, and 50ms, setting M_Delay to 50% cuts those times in half (100ms, 75ms, 50ms, and 25ms)).
m_Fback	Scales all the individual detune delay feedbacks, effectively serving as a master feedback (i.e. if M_Fback is set to 0%, all the delays will have no feedback, regardless of their individual settings. Alternatively, you could also have no feedback with M_Fback set to 100% if you set each delay's feedback parameter to 0%.).
m_Pan	Scales all the individual detune pans, effectively serving as an image spread control (i.e. if M_Pan is set to 0%, all the delays will be panned center, regardless of their individual settings).
Tightnes	Controls the "length" of the detuner. Shorter values yield tighter, albeit glitched results. Longer values yield deglitched, albeit "looser" results. Find the happy medium!

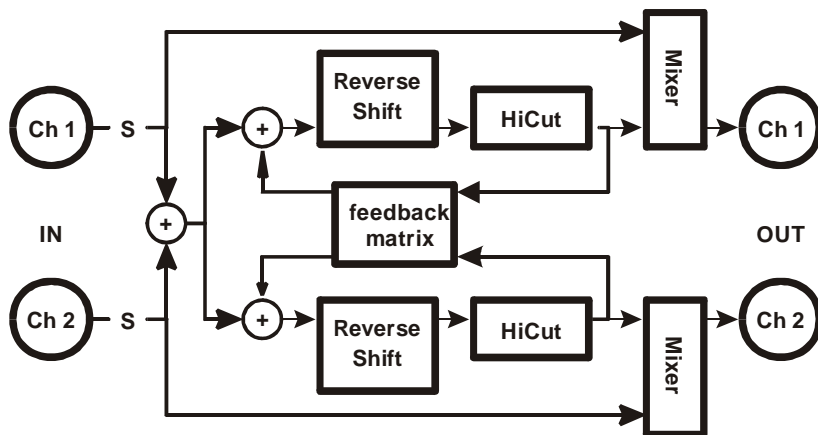| Shift^(x = 1, 2, 3, 4)	
Detune x	Controls the detune amount of detune x.
Level x	Controls the level of detune x.
Delay x	Controls the delay time of detune x.
Fback x	Controls how much of detune x's output is reapplied to its input.
Pan x	Controls where in the stereo field detune x will appear.

reverse crystals

(161   ReverseCrystals)

Here we have two five second reverse shifters where the feedback signal can be the sum of the outputs, the channel output or the other channel for 'crisscross' effects. Note that the delay times are independent of the splice length (keep at 0 ms for splicing referenced to actual delay perceived).

Stereo in, stereo out

Masters	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the input to the loop with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
Mix	Controls the wet-to-dry ratio. If set to "0%", you won't hear any "crystals". If set to "100%" you'll hear nothing but "crystals".
m_Fback	Scales all the individual delay feedbacks, effectively serving as a master feedback (i.e. if M_Fback is set to 0%, all the delays will have no feedback, regardless of their individual settings. Alternatively, you could also have no feedback with M_Fback set to 100% if you set each delay's feedback parameter to 0%.).
FB Type	Determines if the feedback from each "crystal" will return to its own input, to the other delay's input, or to both.
Hicut	Controls the lowpass filter cutoff frequency.

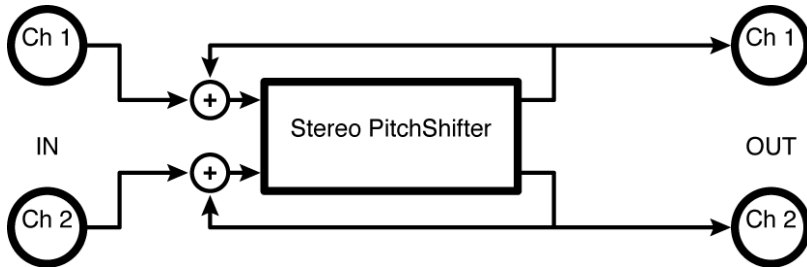| Crystals^(x = 1, 2)	
Pitch x	Controls the pitch of reverse shifter x.
Fback x	Controls how much of reverse shifter x's output is reapplied to its input.
Length x	Controls the audible delay time. Either reflects this time as entered by the T_Length to its right (given the current tempo) or allows you to enter the delay time in milliseconds.
T_Length	Controls the audible delay time. Controls this time in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".
Delay x	Controls the pitchshifting delay (typically very short). Either reflects the physical delay time as entered by T_Delayx (given the current tempo) or allows you to enter the delay time in milliseconds .
T_Delay x	Controls the pitchshifting delay (typically very short). Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".

stereoshift

(162   St Shifter)

This phase-coherent, non-diatonic shifter has simple controls. It is designed to preserve the stereo 'image' of the input signal.
Stereo in, stereo out

Shift

Send   *Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*

Level   *Controls the output level.*

Pitch   *Controls the amount of pitchshift effect.*

Delay   *Controls the delay time for optimal performance.*

Fback   *Controls how much of the pitchshifter's output is reapplied to its input.*

LowNote *Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.*

XFade   *Controls the crossfade between old pitchshift values and new pitchshift values.*

diatonicshift 2

(163   DiatonicShift2)

A single two-voice diatonic shifter whose feedback path is the sum of both outputs, with 'master' and individual parameters. *LFO* and external modulation are available.
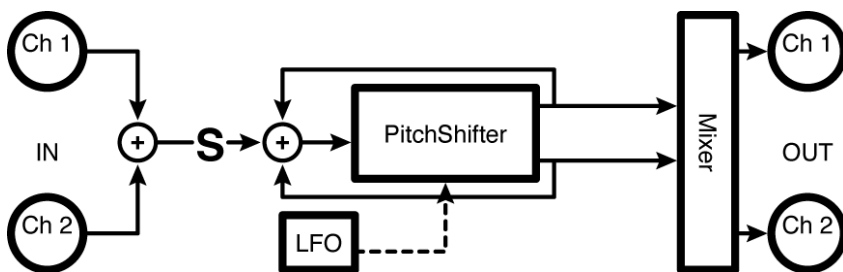*Summed in, stereo out*

Masters

Send   *Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*

m_Level   *Scales both pitchshifter levels, effectively serving as a master volume.*

m_Dly   *Scales both delay times, effectively serving as a master delay.*

m_Mod   *Scales both modulation amounts, effectively serving as a... you guessed it.*

Eclipse Algorithms Manual

m_Pan   *Scales both pitchshifter pans, effectively serving as an image spread control.*

Cal	
Tune	*Controls the detector's calibration, allowing for deviation from A-440 tuning.*
Key	*Selects the key that the detector references (if a scale is in fact selected under Scale).*
Scale	*Selects the scale that the detector references.*
Quant	*Determines the level of "quantization", or how close the Eclipse will bring an "off" note to "perfection".*
LowNote	*Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.*
XFade	*Controls the crossfade between old pitchshift values and new pitchshift values.*
Glide	*Controls how quickly the shifter slides between shift values.*

Shift^(x = 1, 2)	
Pitch x	*Controls the amount of pitchshift effect for pitchshifter x.*
Mod x	*Controls the amount of pitch modulation for a vibrato effect.*
Manual_M	*Allows users without an external pedal to modulate the effect from the front panel.*
Delay x	*Either reflects the physical delay time as entered by T_Delayx (given the current tempo) or allows you to enter the delay time in milliseconds .*
T_Delay x	*Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".*

LFO	
Mode	*Selects whether the LFO will modulate each shifter in phase ("lfo sync") or out of phase ("lfo oppose"). Also allows an external source to modulate the shifters ("manual/ext").*
Rate	*Controls the modulation rate if T_Rate is set to "off" or reflects the physical modulation rate as selected by T_Rate and the system tempo.*
T_Rate	*Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".*
Shape	*Determines the shape of the modulating signal.*
Duty	*Controls the "duty cycle" of the modulating wave for any waveform other than "sine". For example, if you have selected "triangle" under Shape and set Duty to "0%", the triangle will morph to a sawtooth (very fast rise, slow fall). With Duty at "100%", the triangle will morph to a ramp (slow rise, very fast fall).*
Retrig	*Whenever **Rate** is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of **Retrig** controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.*
Angle	*Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.*

Mix^(x = 1, 2)	
Level x	*Controls the output level of pitch shifter x.*
Pan x	*Controls the stereo placement of pitch shifter x.*
Fback x	*Controls how much of pitch shifter x's output is reapplied to its input.*

Eclipse Algorithms Manual

multishift 2

(164   MultiShift2)

A single two-voice, non-diatonic shifter whose feedback path is the sum of both outputs, with 'master' and individual parameters. *LFO* and external modulation are available.
Summed in, stereo out

See "diatonic 2" above for diagram.

Masters	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
m_Level	Scales both pitchshifter levels, effectively serving as a master volume.
m_Pitch	Scales both pitch shift amounts, effectively serving as a master pitch.
m_Mod	Scales both modulation amounts, effectively serving as a master modulation amount.
m_Dly	Scales both delay times, effectively serving as a master delay.
m_Pan	Scales both pitchshifter pans, effectively serving as an image spread control.

	Cal	
	LowNote	Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.
	XFade	Controls the crossfade between old pitchshift values and new pitchshift values.

Shift^(x = 1, 2)	
Pitch x	Controls the amount of pitchshift effect for pitchshifter x.
Mod x	Controls the amount of pitch modulation for a vibrato effect.
Delay x	Either reflects the physical delay time as entered by T_Delayx (given the current tempo) or allows you to enter the delay time in milliseconds .
T_Delay x	Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".
Manual_M	Allows users without an external pedal to modulate the effect from the front panel.

LFO	
Mode	Selects whether the LFO will modulate each shifter in phase ("lfo sync") or out of phase ("lfo oppose"). Also allows an external source to modulate the shifters ("manual/ext").
Rate	Controls the modulation rate if T_Rate is set to "off" or reflects the physical modulation rate as selected by T_Rate and the system tempo.
T_Rate	Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".
Shape	Determines the shape of the modulating signal.
Duty	Controls the "duty cycle" of the modulating wave for any waveform other than "sine". For example, if you have selected "triangle" under Shape and set Duty to "0%", the triangle will morph to a sawtooth (very fast rise, slow fall). With Duty at "100%", the triangle will morph to a ramp (slow rise, very fast fall).
Retrig	Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.
Angle	Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.

Mix^(x = 1, 2)

Level x   *Controls the output level of pitch shifter x.*

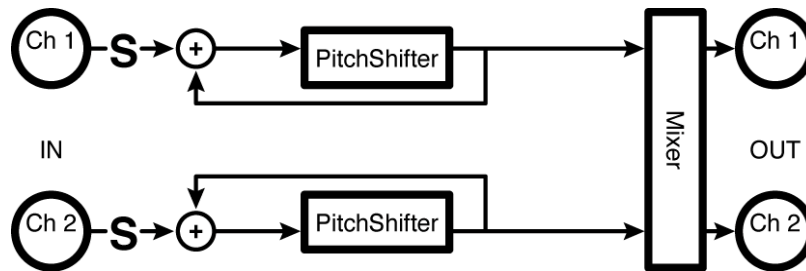**Pan x**   *Controls the stereo placement of pitch shifter x.*

Fback x   *Controls how much of pitch shifter x's output is reapplied to its input.*

---

dual diatonic

(165   DualDiatonic)

Here we have dual two-voice diatonic shifters with individual feedback paths as well as 'master' and individual channel parameters.

Stereo in, stereo out

Masters

Send   *Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*

m_Level   *Scales both pitchshifter levels, effectively serving as a master volume.*

m_Dly   *Scales both delay times, effectively serving as a master delay.*

m_Mod   *Scales both modulation amounts, effectively serving as a... you guessed it.*

m_Pan   *Scales both pitchshifter pans, effectively serving as an image spread control.*

Cal	
Tune	*Controls the detector's calibration, allowing for deviation from A-440 tuning.*
Key	*Selects the key that the detector references (if a scale is in fact selected under Scale).*
Scale	*Selects the scale that the detector references.*
Quant	*Determines the level of "quantization", or how close the Eclipse will bring an "off" note to "perfection".*
LowNote	*Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.*
XFade	*Controls the crossfade between old pitchshift values and new pitchshift values.*
Glide	*Controls how quickly the shifter slides between shift values.*

Shift^(x = 1, 2)

Pitch x   *Controls the amount of pitchshift effect for pitchshifter x.*

Mod x   *Controls the amount of pitch modulation for a vibrato effect.*

Manual_M *Allows users without an external pedal to modulate the effect from the front panel.*

Delay x   *Either reflects the physical delay time as entered by T_Delayx (given the current tempo) or allows you to enter the delay time in milliseconds .*

T_Delay x *Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".*

LFO	
Mode	*Selects whether the LFO will modulate each shifter in phase ("lfo sync") or out of phase ("lfo oppose"). Also allows an external source to modulate the shifters ("manual/ext").*
Rate	*Controls the modulation rate if T_Rate is set to "off" or reflects the physical modulation rate as selected by T_Rate and the system tempo.*
T_Rate	*Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".*
Shape	*Determines the shape of the modulating signal.*
Duty	*Controls the "duty cycle" of the modulating wave for any waveform other than "sine". For example, if you have selected "triangle" under Shape and set Duty to "0%", the triangle will morph to a sawtooth (very fast rise, slow fall). With Duty at "100%", the triangle will morph to a ramp (slow rise, very fast fall).*
Retrig	*Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.*
Angle	*Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.*

Mix^(x = 1, 2)	
Level x	*Controls the output level of pitch shifter x.*
Pan x	*Controls the stereo placement of pitch shifter x.*
Fback x	*Controls how much of pitch shifter x's output is reapplied to its input.*

dual multishift

(166   DualMultishift)

This algorithm offers dual two-voice non-diatonic shifters with individual feedback paths as well as 'master' and individual channel parameters.

Stereo in, stereo out

See "dual diatonic" above for diagram.

Masters	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
m_Level	*Scales both pitchshifter levels, effectively serving as a master volume.*
m_Pitch	*Scales both pitch shift amounts, effectively serving as a master pitch.*
m_Mod	*Scales both modulation amounts, effectively serving as a master modulation amount.*
m_Dly	*Scales both delay times, effectively serving as a master delay.*
m_Pan	*Scales both pitchshifter pans, effectively serving as an image spread control.*

Cal	
LowNote	*Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.*
XFade	*Controls the crossfade between old pitchshift values and new pitchshift values.*

Shift^(x = 1, 2)	
Pitch x	*Controls the amount of pitchshift effect for pitchshifter x.*
Mod x	*Controls the amount of pitch modulation for a vibrato effect.*

Delay x	Either reflects the physical delay time as entered by T_Delayx (given the current tempo) or allows you to enter the delay time in milliseconds .
T_Delay x	Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".
Manual_M	Allows users without an external pedal to modulate the effect from the front panel.

LFO	
Mode	Selects whether the LFO will modulate each shifter in phase ("lfo sync") or out of phase ("lfo oppose"). Also allows an external source to modulate the shifters ("manual/ext").
Rate	Controls the modulation rate if T_Rate is set to "off" or reflects the physical modulation rate as selected by T_Rate and the system tempo.
T_Rate	Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".
Shape	Determines the shape of the modulating signal.
Duty	Controls the "duty cycle" of the modulating wave for any waveform other than "sine". For example, if you have selected "triangle" under Shape and set Duty to "0%", the triangle will morph to a sawtooth (very fast rise, slow fall). With Duty at "100%", the triangle will morph to a ramp (slow rise, very fast fall).
Retrig	Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.
Angle	Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.

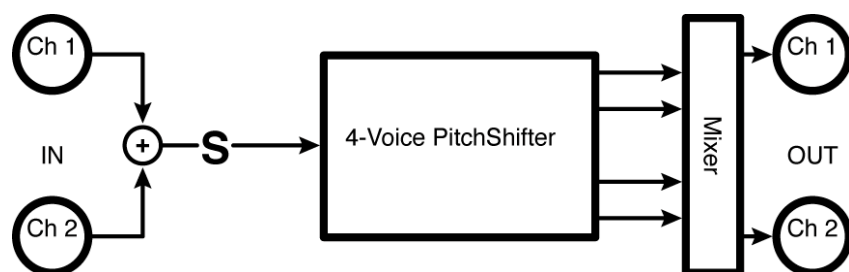| **Mix^(x = 1, 2)**	
Level x	Controls the output level of pitch shifter x.
Pan x	Controls the stereo placement of pitch shifter x.
Fback x	Controls how much of pitch shifter x's output is reapplied to its input.

diatonic 4

(167   DiatonicShift4)

This single four-voice diatonic shifter has 'master' and individual parameters (except 'pitch').
Summed in, stereo out

Masters	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
m_Level	Scales both pitchshifter levels, effectively serving as a master volume.
m_Dly	Scales both delay times, effectively serving as a master delay.
m_Pan	Scales both pitchshifter pans, effectively serving as an image spread control.

Eclipse Algorithms Manual

Cal	
Tune	*Controls the detector's calibration, allowing for deviation from A-440 tuning.*
Key	*Selects the key that the detector references (if a scale is in fact selected under Scale).*
Scale	*Selects the scale that the detector references.*
Quant	*Determines the level of "quantization", or how close the Eclipse will bring an "off" note to "perfection".*
LowNote	*Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.*
XFade	*Controls the crossfade between old pitchshift values and new pitchshift values.*
Glide	*Controls how quickly the shifter slides between shift values.*

Shift^(x = 1, 2, 3, 4)	
Pitchx	*Controls the amount of pitchshift effect for pitchshifter x.*
Modx	*Controls the amount of pitch modulation for a vibrato effect.*
Manual_M	*Allows users without an external pedal to modulate the effect from the front panel.*
Delayx	*Either reflects the physical delay time as entered by T_Delayx (given the current tempo) or allows you to enter the delay time in milliseconds .*
T_Delayx	*Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".*

Mix^(x = 1, 2, 3, 4)	
Level x	*Controls the output level of pitch shifter x.*
Pan x	*Controls the stereo placement of pitch shifter x.*

multishift 4

(168   MultiShift4)

A single four voice non-diatonic shifter with 'master' and individual parameters.
Summed in, stereo out

See "diatonic 4" above for diagram.

Masters	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
m_Level	*Scales both pitchshifter levels, effectively serving as a master volume.*
m_Pitch	*Scales both pitch shift amounts, effectively serving as a master pitch.*
m_Dly	*Scales both delay times, effectively serving as a master delay.*
m_Pan	*Scales both pitchshifter pans, effectively serving as an image spread control.*

Cal	
LowNote	*Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.*
XFade	*Controls the crossfade between old pitchshift values and new pitchshift values.*

Shift^(x = 1, 2)	
Pitch x	*Controls the amount of pitchshift effect for pitchshifter x.*
Delay x	*Either reflects the physical delay time as entered by T_Delayx (given the current tempo) or allows you to enter the delay time in milliseconds .*
T_Delay x	*Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".*

***Manual*_M** *Allows users without an external pedal to modulate the effect from the front panel.*

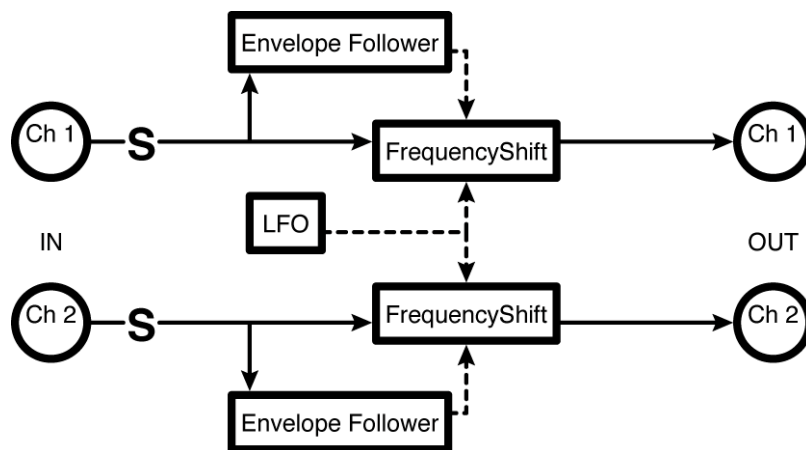| ***Mix^(x = 1, 2)***	
Level x	*Controls the output level of pitch shifter x.*
Pan x	*Controls the stereo placement of pitch shifter x.*

dual modfreqshift

(170   ModFreqShift)

This two-channel algorithm contains sweepable ring modulators, controlled by 'master' and individual channel parameters. The 'mode' selector allows the modulation source to be selected from *LFO*, envelope or pedal.
Stereo in, stereo out

Master	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
Mix	*Determines the wet (ring mod) to dry ratio for both rings (you can overwrite this value under Ring#1 or Ring#2.*
Mode	*Selects whether the LFO, the envelope generator, or an external pedal will modulate both rings (you can overwrite this value under Ring#1 or Ring#2).*
m_Rate	*Scales the individual LFO rates.*
m_Ring	*Scales the individual ring amounts.*
m_Mod	*Scales the individual modulation amounts.*
Man_Ped	*Allows users without an external pedal to modulate the effect from the front panel.*

Env/LFO	
Env	*Determines which input or inputs the envelope follower will track.*
Sens	*Controls how sensitive the envelope follower is; alter this parameter to match the average input level to the amount of modulation desired.*
Attack	*Controls how quickly the envelope follower will track an increase at its input.*
Decay	*Controls how quickly the envelope follower will track a decrease at its input.*
Shape	*Determines the shape of the modulating signal.*

Duty	*Controls the "duty cycle" of the modulating wave for any waveform other than "sine". For example, if you have selected "triangle" under Shape and set Duty to "0%", the triangle will morph to a sawtooth (very fast rise, slow fall). With Duty at "100%", the triangle will morph to a ramp (slow rise, very fast fall).*
Retrig	*Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.*
Angle	*Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.*

Ring#1	
Mix1	*Determines the wet (ring mod) to dry ratio for ring 1 (you can overwrite this value under Master).*
Mode1	*Selects whether the LFO, the envelope generator, or an external pedal will modulate ring 1 (you can overwrite this value under Master).*
Rate1	*Controls the modulation rate for ring 1 if T_Rate1 is set to "off" or reflects the physical modulation rate as selected by T_Rate1 and the system tempo.*
T_Rate	*Controls the modulation rate on ring 1 in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".*
Ring1	*Either reflects the physical ring depth as entered by T_Ring1 (given the current tempo) or allows you to enter the ring depth in Hertz if T_Ring1 is set to "Off".*
T_Ring1	*Controls ring depth 1 in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic ring depth generation "off". The actual ring depth displayed under Ring1 is a function of this control and the current system tempo.*
RMod1	*Controls how much the ring depth varies in response to the modulation source.*
Man_Ped1	*Allows users without an external pedal to modulate the effect from the front panel.*

Ring#2	

Parameters analogous to Ring#1.

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verbs

A diffusor *creates a dense field of delay repeats that is typically used to create* reverberator *structures - it may also be thought of as essentially a chain of series-connected* allpass *filters.*
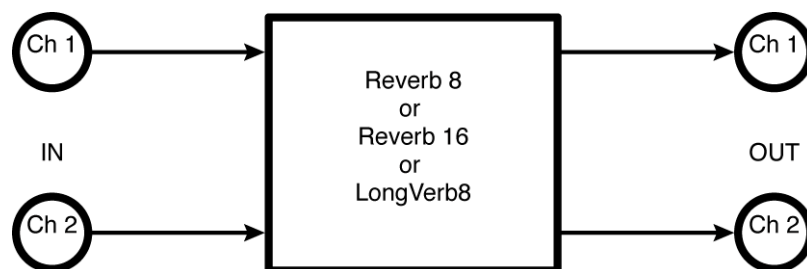*A* reverb *proper usually consists of a* pre-delay, *to give early reflections, followed by a number of reverb* lines, *which are delay sections with multiple feedback loops. Usually, a reverb with more lines will be smoother and more natural sounding.*

---

long reverb 8

(175   LongVerb8)

This reverb has a long decay with eight two-second lines. Decay time is controlled by tempo (*rdecay*).
Stereo in, stereo out

Verb	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
R Level	*Controls the output level.*
RDecay	*Either reflects the physical reverb decay as entered by T_RDecay (given the current tempo) or allows you to enter the reverb decay in seconds if T_RDecay is set to "Off".*
T_RDecay	*Controls the reverb decay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic decay generation "off".*
RSize	*Controls the reverb's room size by scaling the actual delay times.*
PreDly	*Controls how long the predelay is. Predelay is inserted before the reverb because in a real room, reverb arrives at a listener's ears some time after the dry sound. The larger the room, the longer the lag.*
HiCut	*Controls the cut of a high-frequency shelving EQ.*
LowCut	*Controls the cut of a low-frequency shelving EQ.*
HiFreq	*Controls the frequency at which the high-frequency shelving EQ begins to act.*
LowFreq	*Controls the frequency at which the low-frequency shelving EQ begins to act.*
Depth	*Controls how much individual delay times will vary in response to the LFO controlled by Rate. Determines the "thickness" of the chorus effect.*
Rate	*Scales each delay's LFO rate, effectively serving as a master rate.*
Span	*Scales each delay's pan setting, effectively controlling the image width.*
Glide	*Controls how quickly delay times arrive at new values when individual delay times are adjusted (either directly or via RSize).*

reverb 8

(174   Reverb 8)

A short reverb with eight 200-mS lines. Decay time is controlled by tempo.

Stereo in, stereo out

See "long reverb 8" above for diagram and parameters.

reverb 16

(176   Reverb 16)

A dense reverb with sixteen 200-mS lines. Decay time is controlled by tempo.
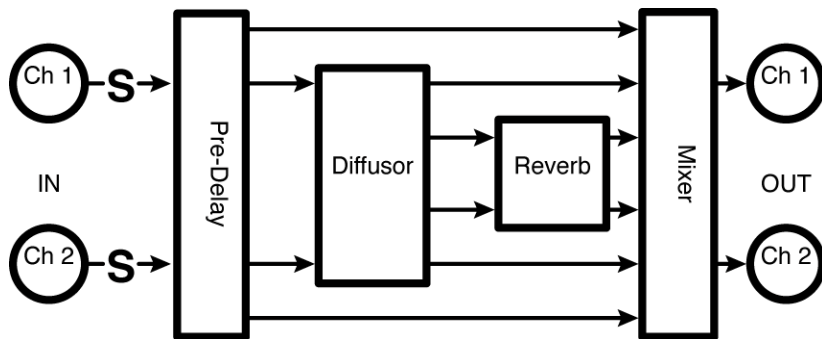
Stereo in, stereo out

See "long reverb 8" above for diagram and parameters.

dense room 16

(178   DenseRoom16)

This dense reverb offers a stereo 200-mS predelay to give early reflection signals. The predelay also feeds two four-part 500-mS diffusors which in turn feed a reverb with sixteen 200-ms lines. (This is an extension of the popular Eventide H3000 structure.) The final out put is a mix of the predelay signals (*early*), the diffusor output (*position*) and the reverb output. Decay time is controlled by tempo.

Stereo in, stereo out

Verb	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
R Level	*Controls the output level.*
Early	*Controls the level of the pre-delay signal, utilized as an early reflection (in a real room, the first, relatively coherent reflection off of the nearest surface).*
Position	*Controls the apparent position in the "room", or how deep you are into the reverberant field.*
Diffusion	*A master feedback control for the diffusors, controlling their "fuzziness" or "distinctness."*
DSize	*Controls the diffusor's "size" by scaling its delay times.*
RDecay	*Either reflects the physical reverb decay as entered by T_RDecay (given the current tempo) or allows you to enter the reverb decay in seconds if T_RDecay is set to "Off".*

T_RDecay	*Controls the reverb decay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic decay generation "off".*
RSize	*Controls the reverb's room size by scaling the actual delay times.*
PreDly	*Controls how long the predelay is. Predelay is inserted before the reverb because in a real room, reverb arrives at a listener's ears some time after the dry sound. The larger the room, the longer the lag.*
HiCut	*Controls the cut of a high-frequency shelving EQ.*
LowCut	*Controls the cut of a low-frequency shelving EQ.*
HiFreq	*Controls the frequency at which the high-frequency shelving EQ begins to act.*
LowFreq	*Controls the frequency at which the low-frequency shelving EQ begins to act.*
Depth	*Controls how much individual delay times will vary in response to the LFO controlled by Rate. Determines the "thickness" of the chorus effect.*
Rate	*Scales each delay's LFO rate, effectively serving as a master rate.*
Span	*Scales each delay's pan setting, effectively controlling the image width.*
Glide	*Controls how quickly delay times arrive at new values when individual delay times are adjusted (either directly or via RSize).*

dense room 8

(177   DenseRoom 8)

This reverb is similar to "dense room 16," above, with only eight reverb lines.
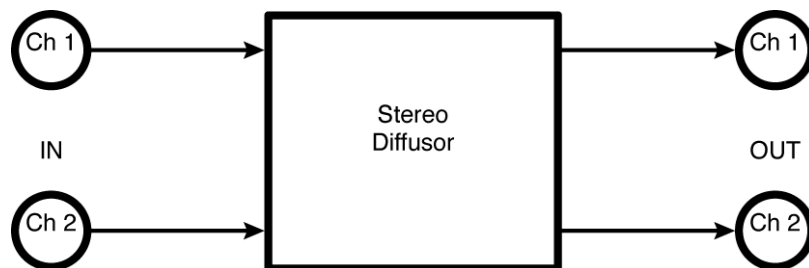
Stereo in, stereo out

See "dense room 16" above for diagram and parameters.

s_diffusor

(180   St Diffusor)

This algorithm contains two four-part 500-mS diffusors.

Stereo in, stereo out

Verb

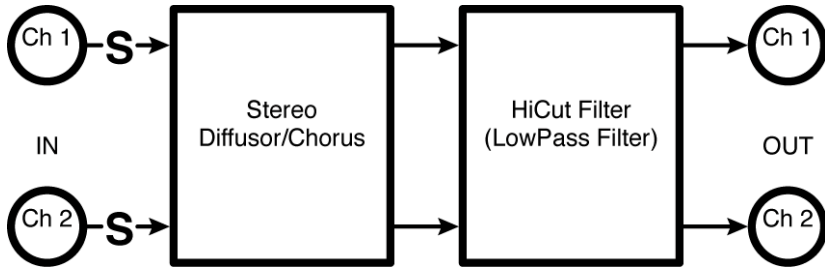|	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
Level	*Controls the output level.*
Diffusion	*A master feedback control for the diffusors, controlling their "fuzziness" or "distinctness."*
DSize	*Controls the diffusor's "size" by scaling its delay times.*

Eclipse Algorithms Manual

diffchorus

(181   Diffchorus)

This algorithm contains two four-part 500-mS diffusors with internal modulation to randomize the effect. An adjustable high-cut filter may be used to smooth out the result.
Stereo in, stereo out

Diffusor

Send   Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.

Level   Controls the output level.

Diffusion   A master feedback control for the diffusors, controlling their "fuzziness" or "distinctness."

DSize   Controls the diffusor's "size" by scaling its delay times.

Depth   Controls how much individual delay times will vary in response to the LFO controlled by Rate. Determines the "thickness" of the chorus effect.

Rate   Scales each delay's LFO rate, effectively serving as a master rate.

HiCut   Controls the cutoff frequency of a gentle low-pass filter.

Span   Scales each delay's pan setting, effectively controlling the image width.

Glide   Controls how quickly delay times arrive at new values when individual delay times are adjusted (either directly or via RSize).
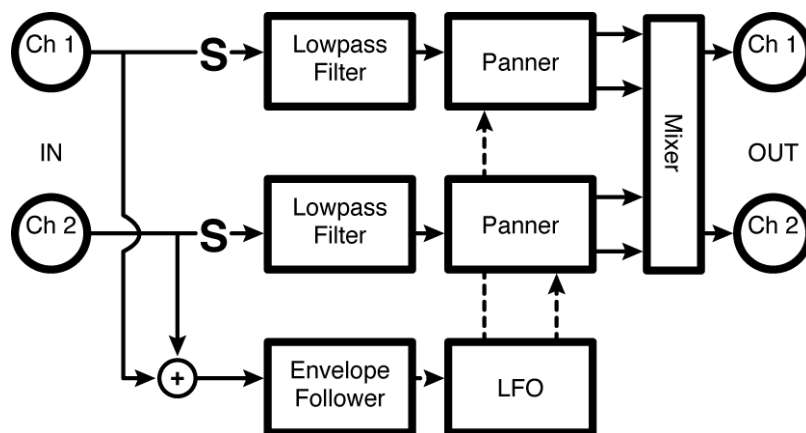
combos

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fm modfilter/pan

(152   FmFilterPan)

This algorithm consists of a modulated filter feeding a panner. The modulation source for both the filter and the panner is controlled by the envelope of the input.
Stereo in, stereo out

Filt/Pan	
Send	Adjusts the level at 'S' in the diagram. If you want to modulate the input to the loop with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.
Level	Controls the output level of the filter..
Freq	Controls the "baseline" center frequency or cutoff frequency.
Fmod	Controls how far from the "baseline" the cutoff frequency will deviate in response to the modulation signal.
Q	Controls the "baseline" bandwidth or resonance.
Qmod	Controls how far from the "baseline" the Q will deviate in response to the modulation signal.
Pan	Determines if the modulation will work in phase ("sync") or out of phase ("oppose").

LFO	
Rate	Controls the modulation rate if T_Rate1 is set to "off" or reflects the physical modulation rate as selected by T_Rate and the system tempo.
T_Rate	Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".
FMRate	Analogous to Rate, except that this pertains to modulation of the LFO's signal (not to modulation of the panner directly).
T_FMRate	Analogous to T_Rate, except that this pertains to modulation of the LFO's signal (not to modulation of the panner directly).
Shape	Determines the shape of the modulating signal.
Duty	Controls the "duty cycle" of the modulating wave for any waveform other than "sine". For example, if you have selected "triangle" under Shape and set Duty to "0%", the triangle will morph to a sawtooth (very fast rise, slow fall). With Duty at "100%", the triangle will morph to a ramp (slow rise, very fast fall).

Eclipse Algorithms Manual

Polarity   When set to "unipolar", modulations add the value of Fmod to Freq. When set to "bipolar", modulations add the value of Fmod to Freq and subtract the value of Fmod from Freq (the same goes for Q and Qmod).

Retrig   Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.

Angle   Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.
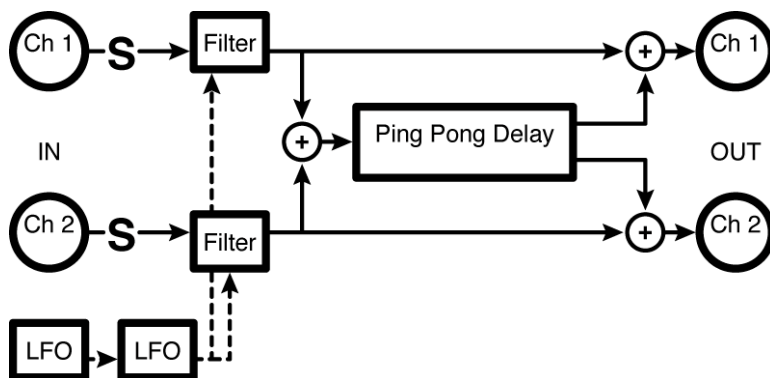
Env

Sens   Controls how sensitive the envelope follower is; alter this parameter to change the "depth" of the FM effect.

Attack   Controls how fast the envelope follower will track an increase in volume.

Decay   Controls how fast the envelope follower will track a decrease in volume.

LFO filter+pingpong

(153   LfoFilter+Pong)

This effect consists of two filters modulated by two *LFO*s. The first LFO controls the rate of modulation, while the second controls the depth of that modulation, giving a sort of vintage synth effect. The stereo output of these filters is then mixed with the dry input signal, and summed to feed a ping-pong delay with pre-delay. Tempo controls rates and delay times.
Stereo in, stereo out

Mod^(x = 1, 2)

Rate x   Controls LFO x's modulation rate if T_Ratex is set to "off" or reflects the physical LFO modulation rate as selected by T_Ratex and the system tempo.

T_Rate x   Controls LFO x's modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic LFO modulation generation "off".

Wave x   Determines the shape of filter x's modulating signal.

Retrig   Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.

Angle   Given most configurations of the Eclipse, the LFOs will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in their waveform the LFOs begin.

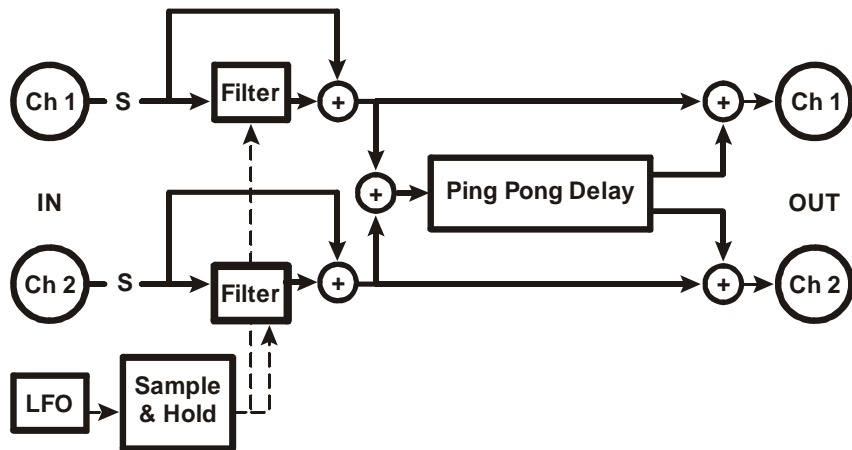| *Filter*	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
Mix	*Controls the ratio between the dry and wet (filtered) signal.*
Freq	*Controls the "baseline" cutoff frequency.*
Fmod	*Controls how far from the "baseline" the cutoff frequency will deviate in response to the modulation signal.*
Q	*Controls the "baseline" resonance.*
Qmod	*Controls how far from the "baseline" the Q will deviate in response to the modulation signal.*

Delays	
Level	*Controls the output level of the ping pong delay.*
PDelay	*Either reflects the physical pre-delay time as entered by T_PDelay (given the current tempo) or allows you to enter the pre-delay time in milliseconds if T_PDelay is set to "Off".*
T_PDelay	*Controls the pre-delay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic pre-delay generation "off".*
Pong	*Either reflects the physical pong time as entered by T_Pong (given the current tempo) or allows you to enter the pong time in milliseconds if T_Pong is set to "Off".*
T_Pong	*Controls the pong in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic pong generation "off". The actual pong time displayed under Pong is a function of this control and the current system tempo.*
FBack	*Controls how much of the pong's output is reapplied to its input.*

s/h filter+pingpong

(154   S/HFilter+Pong)

The output of stereo sample-and-hold filters is mixed with the dry input signal, and are also summed to feed a ping-pong delay with pre-delay. Tempo controls rates and delay times.
Stereo in, stereo out

S/H	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the input to the loop with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
Mix	*Controls the ratio between the dry and wet (S/H) signal.*
***F* Min**	*Sets the minimum frequency for the Sample and Hold.*

Eclipse Algorithms Manual

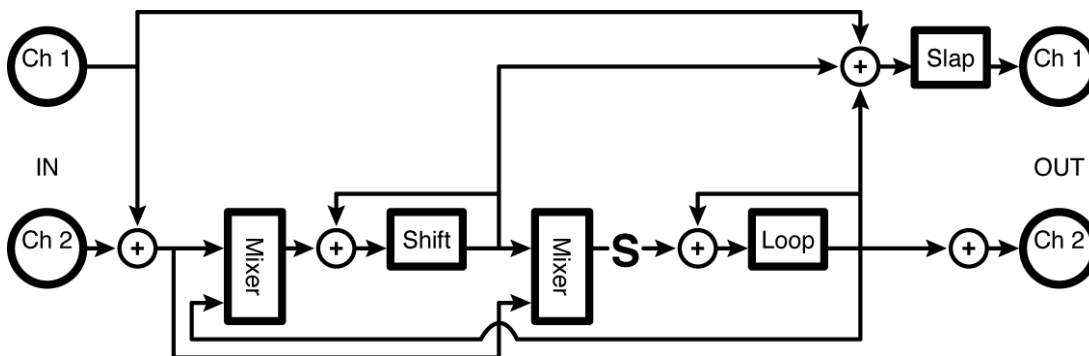|	
F Max	*Sets the maximum frequency for the Sample and Hold.*
Reson	*Controls the filter resonance.*
Flux	*Controls how quickly the filter moves between "notes".*
Rate	*Controls the modulation rate if T_Rate is set to "off" or reflects the physical modulation rate as selected by T_Rate and the system tempo.*
T_Rate	*Controls the modulation rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic modulation generation "off".*
Retrig	*Whenever Rate is changed (either manually or by updating the tempo), the new Rate is crossfaded with the old Rate. The value of Retrig controls how long the crossfade is as a percentage of the old Rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old Rate.*
Angle	*Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.*

Delays

Level	*Controls the output level of the ping pong delay.*
PDelay	*Either reflects the physical pre-delay time as entered by T_PDelay (given the current tempo) or allows you to enter the pre-delay time in milliseconds if T_PDelay is set to "Off".*
T_PDelay	*Controls the pre-delay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic pre-delay generation "off".*
Pong	*Either reflects the physical pong time as entered by T_Pong (given the current tempo) or allows you to enter the pong time in milliseconds if T_Pong is set to "Off".*
T_Pong	*Controls the pong in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic pong generation "off".*
FBack	*Controls how much of the pong's output is reapplied to its input.*

manifold alpha

(155   ManifoldAlpha)

This is a pitch-shifting loop and slap delay. A mixer at the shifter input selects between the source and the loop signals, while another mixer at the loop input selects between the source and the shifted signals. Both outputs (pitch and loop) are then mixed with the left input (channel #1) and fed to the slap delay output on the left side, while the loop and pitch also get fed directly to the right output. Tempo may be used to control delay as well as loop and slap length.
Summed in, stereo out

Shift

S InMix	*Controls what percentage of the pitchshifter's input comes from the right input ("0%") and what percentage comes from the loop output ("100%").*

Eclipse Algorithms Manual

S **Input**	*Controls the input level to the pitchshifter.*
S **Level**	*Controls the output level of the pitchshifter.*
Pitch	*Controls the amount of pitchshifting.*
Delay	*Either reflects the pitchshifter's physical delay time as entered by T_Delay (given the current tempo) or allows you to enter the delay time in milliseconds if T_Delay is set to "Off".*
T_Delay	*Controls the pitchshifter delay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".*
S **Fback**	*Controls how much of the pitchshifter's output is reapplied to its input.*
LowNote	*Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.*
Delay	*Controls the crossfade between old pitchshift values and new pitchshift values.*

Loop	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the input to the loop with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
L **InMix**	*Controls what percentage of the loop's input comes from the right input ("0%") and what percentage comes from the pitchshifter output ("100%").*
L **Input**	*Controls the input level to the loop.*
L **Level**	*Controls the output level of the loop.*
Loop	*Either reflects the physical loop time as entered by #Bars (given the current tempo) or allows you to enter the loop time in milliseconds if #Bars is left untouched and the system tempo is not updated (it's better to use #Bars, OK?).*
#Bars	*Controls the length of the loop in terms of bars. The actual loop time displayed under Loop is a function of this control and the current system tempo.*
L **Fback**	*Controls how much of the loop's output is reapplied to its input.*

Slap	
D **Level**	*Controls the output level of the slap delay.*
Delay	*Either reflects the physical delay time as entered by T_Delay (given the current tempo) or allows you to enter the delay time in milliseconds if T_Delay is set to "Off".*
T_Delay	*Controls the delay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off". The actual delay time displayed under Delay is a function of this control and the current system tempo.*

manifold beta

(156   ManifoldBeta)

This algorithm is essentially a version of "manifold alpha," but it uses a reverse shifter rather than the normal (forward) pitch shifter.
Summed in, stereo out

See "manifold alpha" above for diagram.

Reverse	
S **InMix**	*Controls what percentage of the pitchshifter's input comes from the right input ("0%") and what percentage comes from the loop output ("100%").*
S **Input**	*Controls the input level to the pitchshifter.*
S **Level**	*Controls the output level of the pitchshifter.*
Pitch	*Controls the amount of pitchshifting.*

Length	*Either reflects the pitchshifter's physical delay time as entered by T_Delay (given the current tempo) or allows you to enter the delay time in milliseconds if T_Delay is set to "Off".*
T_Length	*Controls the pitchshifter delay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".*
Delay	*Controls the pitchshifting delay (typically very short).*
S Fback	*Controls how much of the pitchshifter's output is reapplied to its input.*
LowNote	*Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.*
Delay	*Controls the crossfade between old pitchshift values and new pitchshift values.*

Loop	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the input to the loop with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
L InMix	*Controls what percentage of the loop's input comes from the right input ("0%") and what percentage comes from the pitchshifter output ("100%").*
L Input	*Controls the input level to the loop.*
L Level	*Controls the output level of the loop.*
Loop	*Either reflects the physical loop time as entered by #Bars (given the current tempo) or allows you to enter the loop time in milliseconds if #Bars is left untouched and the system tempo is not updated (it's better to use #Bars, OK?).*
#Bars	*Controls the length of the loop in terms of bars. The actual loop time displayed under Loop is a function of this control and the current system tempo.*
L Fback	*Controls how much of the loop's output is reapplied to its input.*

Slap	
D Level	*Controls the output level of the slap delay.*
Delay	*Either reflects the physical delay time as entered by T_Delay (given the current tempo) or allows you to enter the delay time in milliseconds if T_Delay is set to "Off".*
T_Delay	*Controls the delay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".*

dual comp/de-ess

(157   VocalProcess)

Here we have dual compressors feeding filters which then feed further compressors. It is set up as compressors into de-essers, and has 'master' as well as individual parameters.

Stereo in, stereo out

Eclipse Algorithms Manual

Master		

	M_Comp	
Thresh		*Sets the input threshold above which compression of the signal takes place and below which the signal is left alone.*
S Knee		*Controls the width of the "soft knee". The soft knee is a region above the threshold over which the ratio transitions from 1:1 to the selected Ratio. It smoothes out the compression.*
Ratio		*Controls the input to output ratio for signals above the threshold. For example, "6:1" produces 1dB of output for every 6dB of input.*
Gain		*Adjusts the output level to improve gain structure.*
Attack		*Controls how quickly the compressor "clamps down" on a signal after it exceeds the threshold.*
Decay		*Controls how quickly the compressor "lets go" of a signal once it falls below the threshold.*
GR 1		*Displays how much gain reduction (compression) is taking place after delay 1.*
GR 2		*Displays how much gain reduction (compression) is taking place after delay 2.*

	M_Filter	
Freq		*Controls the center frequency of the filter used for frequency-conscious compression.*
Q		*Controls the bandwidth of the filter.*
Type		*Selects the type of filter used for frequency-conscious compression (this is usually set to "bandpass").*

	M_D/S C^These controls affect the frequency-conscious compressor	
D Thresh		*Sets the input threshold above which frequency-conscious compression of the signal takes place and below which the signal is left alone.*
D S Knee		*Controls the width of the "soft knee". The soft knee is a region above the threshold over which the ratio transitions from 1:1 to the selected Ratio. It smoothes out the compression.*
D Ratio		*Controls the input to output ratio for signals above the threshold. For example, "6:1" produces 1dB of output for every 6dB of input.*
D Gain		*Adjusts the output level to improve gain structure.*
D Attack		*Controls how quickly the frequency-conscious compressor "clamps down" on a signal after it exceeds the threshold.*
D Decay		*Controls how quickly the frequency-conscious compressor "lets go" of a signal once it falls below the threshold.*
D/S 1		*Displays how much frequency-conscious gain reduction (compression) is taking place after delay 1.*
D/S 2		*Displays how much frequency-conscious gain reduction (compression) is taking place after delay 2.*

Chan#1		

Parameters similar to those described above, dedicated to compressor 1; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

Chan#2		

Parameters similar to those described above, dedicated to compressor 2; changes made here can be overwritten by those made under Master and vice versa (the most recently entered value "wins")

moddelays+verb8

(183   Chordlys+Verb8)

Four delays feed into an eight-line reverb. The tempo affects the delay times, sweep rate and reverb decay time (*rdecay*).

Stereo in, stereo out

M_Delays	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
D *Level*	*Controls the dry delay output level (but does not affect the delay level sent to the reverb).*
*Dly>***Verb**	*Controls the delay level sent to the reverb (but does not affect the dry delay output level).*
m_Delay	*Scales all the individual delay times, effectively serving as a master delay (e.g. if the four delays are set to 200ms, 150ms, 100ms, and 50ms, setting M_Delay to 50% cuts those times in half (100ms, 75ms, 50ms, and 25ms)).*
m_Depth	*Scales all the individual delay sweep depths, effectively serving as a master depth.*
m_Fback	*Scales all the individual delay feedbacks, effectively serving as a master feedback.*

Delays^(x = 1, 2, 3, 4)	
Delay x	*Either reflects the physical delay time as entered by T_Delayx (given the current tempo) or allows you to enter the delay time in milliseconds .*
T_Delay *x*	*Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".*
Fback x	*Controls how much of delay x's output is reapplied to its input.*

Mods^(x = 1, 2, 3, 4)	
Depth x	*Controls the sweep depth of delay x.*
Rate	*Controls the sweep rate if T_Rate is set to "off" or reflects the physical sweep rate as selected by T_Rate and the system tempo.*
T_Rate	*Controls the sweep rate in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic sweep generation "off". The actual sweep rate displayed under Rate is a function of this control and the current system tempo.*
Shape	*Determines the shape of the modulating signal.*
Retrig	*Whenever the sweep rate is changed (either manually or by updating the tempo), the new rate is crossfaded with the old rate. The value of **Retrig** controls how long the crossfade is as a percentage of the old rate. Thus, a value of 100% means that the crossfade will be complete in one cycle of the old rate.*
Angle	*Given most configurations of the Eclipse, the modulation will "start over" when the tempo button is tapped or when an external sequence is started. Angle determines where in its waveform the modulation begins.*

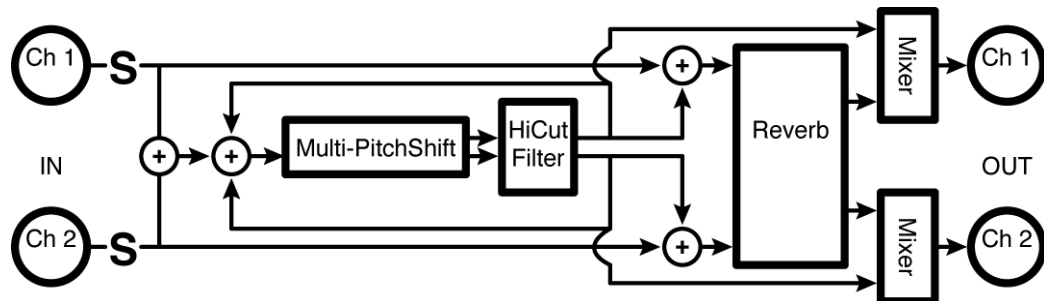| *Verb*	
R *Level*	*Controls the reverb output level.*
RDecay	*Either reflects the physical reverb decay as entered by T_RDecay (given the current tempo) or allows you to enter the reverb decay in seconds if T_RDecay is set to "Off".*
T_RDecay	*Controls the reverb decay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic decay generation "off".*
RSize	*Controls the reverb's room size by scaling the actual delay times.*
PreDly	*Controls how long the predelay is. Predelay is inserted before the reverb because in a real room, reverb arrives at a listener's ears some time after the dry sound. The larger the room, the longer the lag.*
HiCut	*Controls the cut of a high-frequency shelving EQ.*
LowCut	*Controls the cut of a low-frequency shelving EQ.*

HiFreq   *Controls the frequency at which the high-frequency shelving EQ begins to act.*

LowFreq   *Controls the frequency at which the low-frequency shelving EQ begins to act.*

Depth   *Controls how much individual delay times will vary in response to the LFO controlled by Rate. Determines the "thickness" of the chorus effect.*

Rate   *Scales each delay's LFO rate, effectively serving as a master rate.*

Span   *Scales each delay's pan setting, effectively controlling the image width.*

Glide   *Controls how quickly delay times arrive at new values when individual delay times are adjusted (either directly or via RSize).*

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multishift+verb8

(184   Mshift+Reverb8)

In this combination effect, a two-voice shifter feeds a reverb, with mixers at the reverb input and the final outputs. Tempo controls delay times and reverb decay times.
Stereo in, stereo out

Masters	
Send	*Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*
S Level	*Controls the dry pitchshifter output level (but does not affect the pitchshifter level sent to the reverb).*
Mix>Verb	*Controls the pitchshifter level sent to the reverb (but does not affect the dry pitchshifter output level).*
S HiCut	*Controls the cutoff frequency of a lowpass filter placed after the pitchshifter.*

Cal	
LowNote	*Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.*
XFade	*Controls the crossfade between old pitchshift values and new pitchshift values.*

Shift^(x = 1, 2)	
Pitch x	*Controls the amount of pitchshift effect for pitchshifter x.*
Delay x	*Either reflects the physical delay time as entered by T_Delayx (given the current tempo) or allows you to enter the delay time in milliseconds .*
T_Delay x	*Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".*
Fback x	*Controls how much of pitch shifter x's output is reapplied to its input.*
Mod x	*Controls the amount of pitch modulation for a vibrato effect.*
Man_Mod x	*Allows users without an external pedal to modulate the effect from the front panel.*

Verb	
R Level	*Controls the reverb output level.*

Eclipse Algorithms Manual

RDecay   *Either reflects the physical reverb decay as entered by T_RDecay (given the current tempo) or allows you to enter the reverb decay in seconds if T_RDecay is set to "Off".*

T_RDecay   *Controls the reverb decay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic decay generation "off".*

RSize   *Controls the reverb's room size by scaling the actual delay times.*

PreDly   *Controls how long the predelay is. Predelay is inserted before the reverb because in a real room, reverb arrives at a listener's ears some time after the dry sound. The larger the room, the longer the lag.*

HiCut   *Controls the cut of a high-frequency shelving EQ.*

LowCut   *Controls the cut of a low-frequency shelving EQ.*

HiFreq   *Controls the frequency at which the high-frequency shelving EQ begins to act.*

LowFreq   *Controls the frequency at which the low-frequency shelving EQ begins to act.*

Depth   *Controls how much individual delay times will vary in response to the LFO controlled by Rate. Determines the "thickness" of the chorus effect.*

Rate   *Scales each delay's LFO rate, effectively serving as a master rate.*

Span   *Scales each delay's pan setting, effectively controlling the image width.*

Glide   *Controls how quickly delay times arrive at new values when individual delay times are adjusted (either directly or via RSize).*

diatonic+verb8

(185   Dshift+Reverb8)

This algorithm is similar to "multishift+verb8," differing in that a two-voice diatonic shifter replaces the normal shifter.

Stereo in, stereo out

See "multishift+verb8" above for diagram.

Masters

Send   *Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*

S Level   *Controls the dry pitchshifter output level (but does not affect the pitchshifter level sent to the reverb).*

Mix>Verb   *Controls the pitchshifter level sent to the reverb (but does not affect the dry pitchshifter output level).*

Cal

Tune   *Controls the detector's calibration, allowing for deviation from A-440 tuning.*

Key   *Selects the key that the detector references (if a scale is in fact selected under Scale).*

Scale   *Selects the scale that the detector references.*

Quant   *Determines the level of "quantization", or how close the Eclipse will bring an "off" note to "perfection".*

LowNote   *Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.*

XFade   *Controls the crossfade between old pitchshift values and new pitchshift values.*

Glide   *Controls how quickly the shifter slides between shift values.*

Shift^(x = 1, 2)

Pitch x   *Controls the amount of pitchshift effect for pitchshifter x.*

Delay x   *Either reflects the physical delay time as entered by T_Delayx (given the current tempo) or allows you to enter the delay time in milliseconds .*

***T_Delay* x**   *Controls delay x in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic delay generation "off".*

Mod x   *Controls the amount of pitch modulation for a vibrato effect.*

***Man_Mod* x** *Allows users without an external pedal to modulate the effect from the front panel.*

Verb

***R* Level**   *Controls the reverb output level.*

RDecay   *Either reflects the physical reverb decay as entered by T_RDecay (given the current tempo) or allows you to enter the reverb decay in seconds if T_RDecay is set to "Off".*

T_RDecay   *Controls the reverb decay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic decay generation "off".*

RSize   *Controls the reverb's room size by scaling the actual delay times.*

PreDly   *Controls how long the predelay is. Predelay is inserted before the reverb because in a real room, reverb arrives at a listener's ears some time after the dry sound. The larger the room, the longer the lag.*

HiCut   *Controls the cut of a high-frequency shelving EQ.*

LowCut   *Controls the cut of a low-frequency shelving EQ.*

HiFreq   *Controls the frequency at which the high-frequency shelving EQ begins to act.*

LowFreq   *Controls the frequency at which the low-frequency shelving EQ begins to act.*

Depth   *Controls how much individual delay times will vary in response to the LFO controlled by Rate. Determines the "thickness" of the chorus effect.*

Rate   *Scales each delay's LFO rate, effectively serving as a master rate.*

Span   *Scales each delay's pan setting, effectively controlling the image width.*
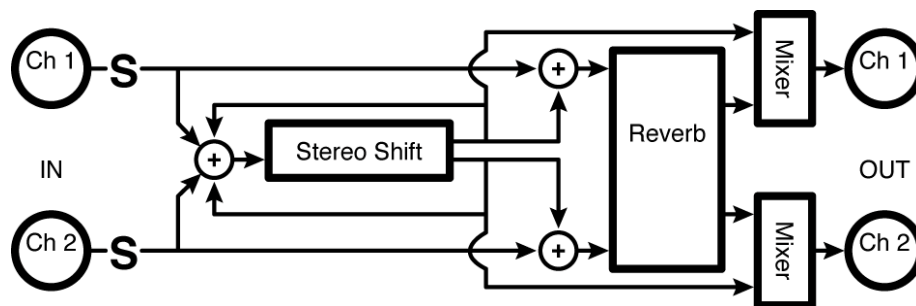
Glide   *Controls how quickly delay times arrive at new values when individual delay times are adjusted (either directly or via RSize).*

stereoshift+verb8

(186   Stshift+Reverb8)

This algorithm is similar to "multishift+verb8," differing in that the shifter is a single voice true stereo unit.

Stereo in, stereo out

Pitch

Send   *Adjusts the level at 'S' in the diagram. If you want to modulate the volume of the effect with MIDI, a volume pedal, or some other external or internal source, this is a fantastic place to do it. You won't get zippering.*

***S* Level**   *Controls the dry pitchshifter output level (but does not affect the pitchshifter level sent to the reverb).*

Mix>Verb   *Controls the pitchshifter level sent to the reverb (but does not affect the dry pitchshifter output level).*

Pitch   *Controls the amount of pitchshift effect.*

Delay	*Controls the pitchshifter delay time.*
Fback	*Controls how much of the pitchshifter's output is reapplied to its input.*
LowNote	*Determines the lowest note the pitchshifter will correctly shift. Set as high as possible for best results.*
XFade	*Controls the crossfade between old pitchshift values and new pitchshift values.*

<div style="border:1px solid black">

Verb

</div>

R Level	*Controls the reverb output level.*
RDecay	*Either reflects the physical reverb decay as entered by T_RDecay (given the current tempo) or allows you to enter the reverb decay in seconds if T_RDecay is set to "Off".*
T_RDecay	*Controls the reverb decay in rhythmic terms (1/4 note, 1/8th note, etc.) or turns rhythmic decay generation "off".*
RSize	*Controls the reverb's room size by scaling the actual delay times.*
PreDly	*Controls how long the predelay is. Predelay is inserted before the reverb because in a real room, reverb arrives at a listener's ears some time after the dry sound. The larger the room, the longer the lag.*
HiCut	*Controls the cut of a high-frequency shelving EQ.*
LowCut	*Controls the cut of a low-frequency shelving EQ.*
HiFreq	*Controls the frequency at which the high-frequency shelving EQ begins to act.*
LowFreq	*Controls the frequency at which the low-frequency shelving EQ begins to act.*
Depth	*Controls how much individual delay times will vary in response to the LFO controlled by Rate. Determines the "thickness" of the chorus effect.*
Rate	*Scales each delay's LFO rate, effectively serving as a master rate.*
Span	*Scales each delay's pan setting, effectively controlling the image width.*
Glide	*Controls how quickly delay times arrive at new values when individual delay times are adjusted (either directly or via RSize).*

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utilities

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mute

(100   Mute_Series, 101 Mute_Xfade)

This is a 'do-nothing' algorithm with no internal connections. Its purpose is to fill the second engine of a single algorithm parallel preset.
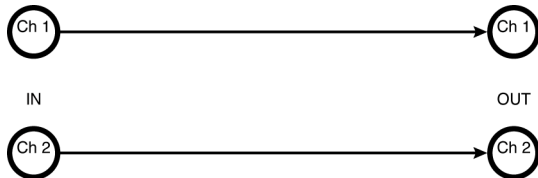*Stereo in, nothing out*

Ch 1 ■   Ch 1

IN   OUT

Ch 2 ■   Ch 2

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thru

(102   Thru_Series, 103 Thru_Xfade)

As the name suggests, the inputs are passed through to the outputs. This is can be useful for troubleshooting and testing, or to fill the second engine of a single algorithm series preset.
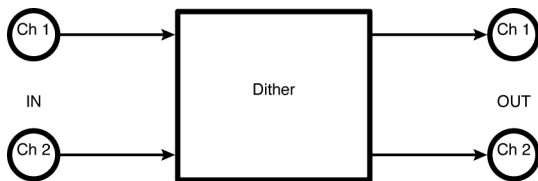*Stereo in, stereo out*

Ch 1 ────────▶ Ch 1

IN   OUT

Ch 2 ────────▶ Ch 2

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dither

(190   Dither)

A simple dithering algorithm that can be used to reduce the number of output bits to 16, 18 or 20 without adding distortion. The user can choose between rectangular (uniform) or triangular distribution.
Stereo in, stereo out

Ch 1 ──▶ [Dither] ──▶ Ch 1

IN   OUT

Ch 2 ──▶ [Dither] ──▶ Ch 2

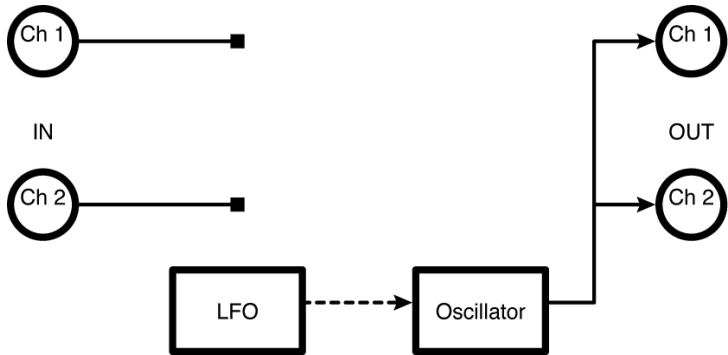| *Operate*	
Bits	*Controls the number of output bits. Match this to the bit depth of the device connected to the Eclipse's digital output.*
Type	*Selects triangle or rectangle dither noise. Stick with triangle unless you know that your audio has already been processed with rectangular dither.*

oscillator (440)

(191   OSC440)

A general purpose sweepable and modulatable oscillator. The default configuration gives a constant (A)440 Hz tone.
Nothing in, summed out

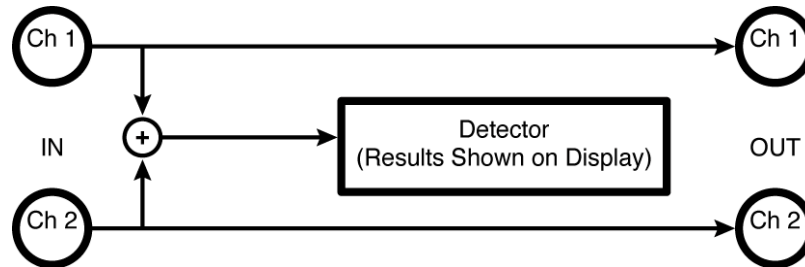| *Osc*	
Level	*Controls the oscillator output level.*
Freq	*Controls the frequency of the oscillator.*
FMod	*Controls how much the frequency modulator (FM) will affect the oscillator.*
Shape	*Selects the waveform of the oscillator.*
Duty	*Controls the "duty cycle" of the oscillator for any Shape other than "sine". For example, if you have selected "triangle" under Shape and set Duty to "0%", the triangle will morph to a sawtooth (very fast rise, slow fall). With Duty at "100%", the triangle will morph to a ramp (slow rise, very fast fall).*
FM Rate	*Controls the rate of the frequency modulator (set it to some multiple of the oscillator's frequency to create interesting timbres).*
FM Shape	*Selects the waveform of the frequency modulator.*
FM Duty	*Controls the frequency modulator's duty cycle.*

Eclipse Algorithms Manual

chromatic tuner

(192   Tuner)

This useful chromatic tuner has sharp/flat metering as well as a display of error amount. It displays pitch (from middle C) and the true frequency. The inputs are passed through unprocessed to the outputs. *Stereo in, stereo out*

Detector	
-|+	*Displays how far from in tune the input is.*
Note	*Displays the closest note to that being supplied at the input.*
OutBy	*Displays how far from in tune the input is (in cents).*
Pitch	*Displays how far the input is from middle C.*
Freq	*Displays the root frequency of the input.*
Cal	*Calibrates the tuner for music that doesn't conform to A-440.*

Cal^(Calibrates the detector for optimal results)	
LowNote	*Determines the lowest note the detector will correctly identify. Set as high as possible for best results.*
HighNote	*Determines the highest note the detector will correctly identify. Set as low as possible for best results.*
Pitch	*Displays how far the input is from middle C.*
Freq	*Displays the root frequency of the input.*