

Filters and Delays

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Basic Filters

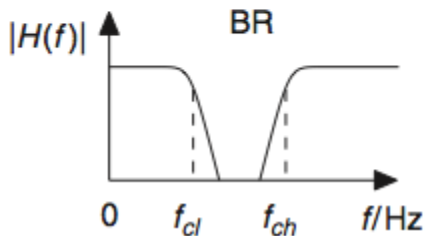
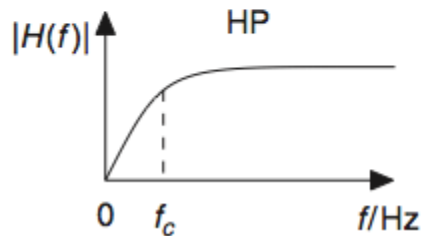
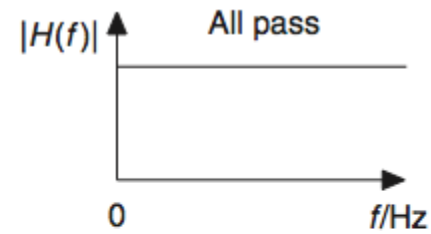
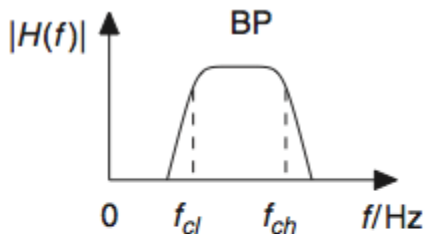
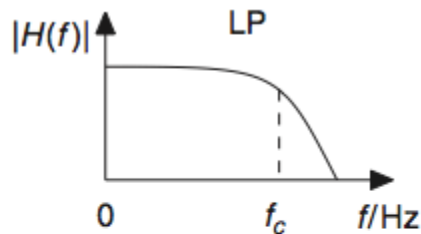
Lowpass (LP)

Highpass (HP)

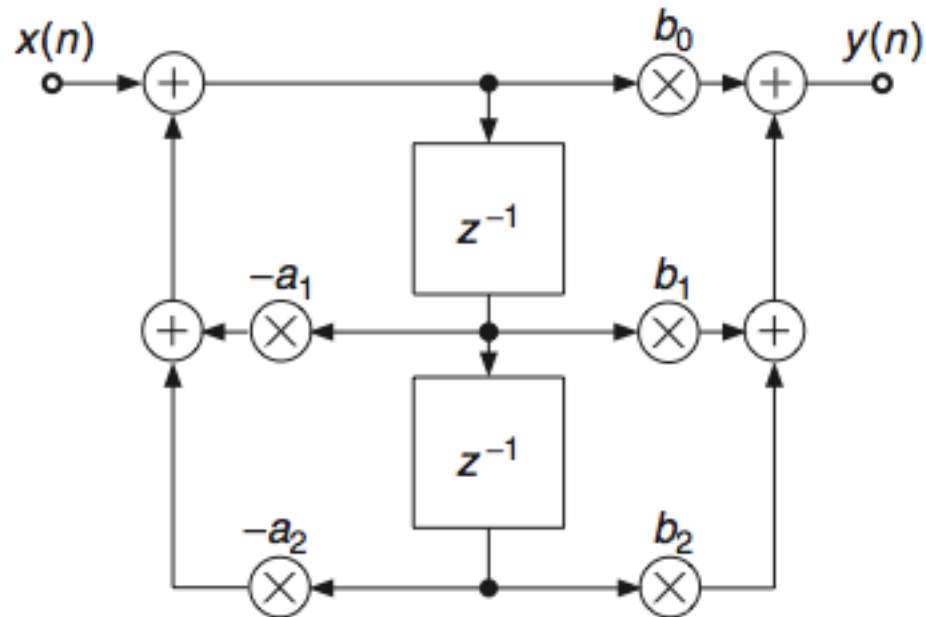
Bandpass (BP)

Bandreject (BR)

Allpass



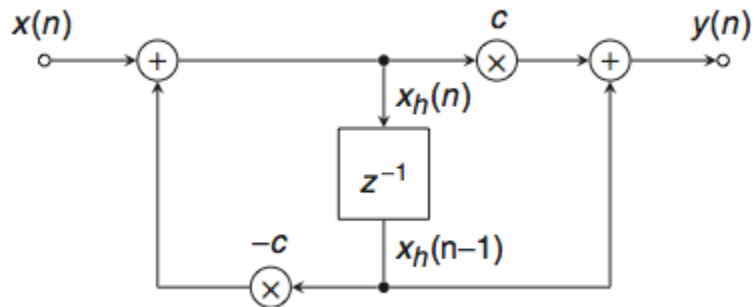
Canonical Filters



	b_0	b_1	a_1
Lowpass	$K/(K+1)$	$K/(K+1)$	$(K-1)/(K+1)$
Highpass	$1/(K+1)$	$-1/(K+1)$	$(K-1)/(K+1)$
Allpass	$(K-1)/(K+1)$	1	$(K-1)/(K+1)$

$$K = \tan(\pi f_c / f_s).$$

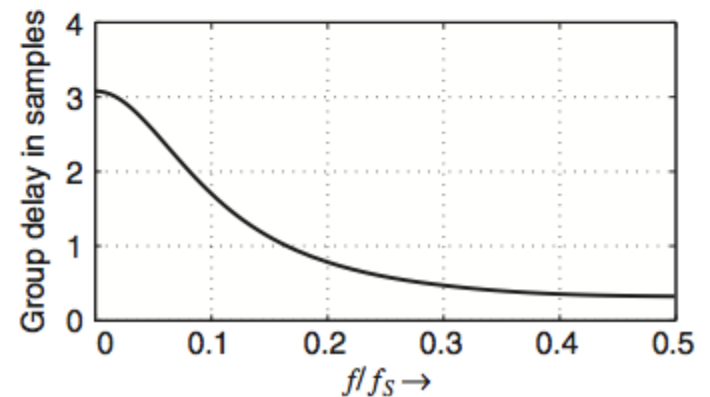
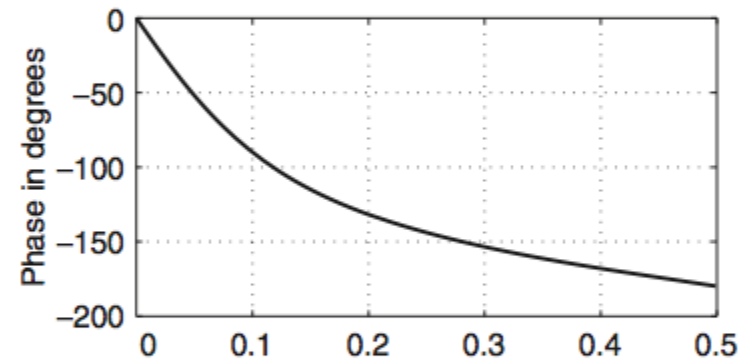
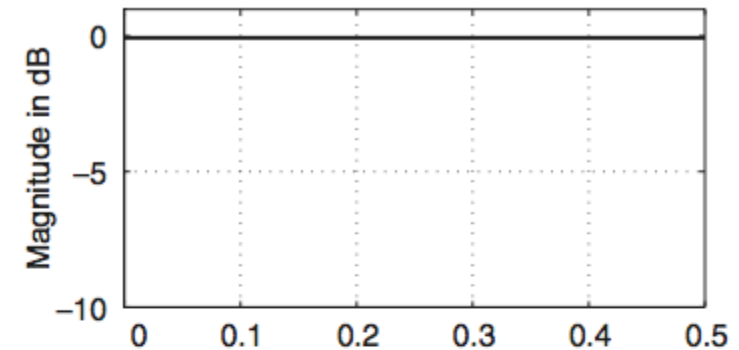
Allpass Filters



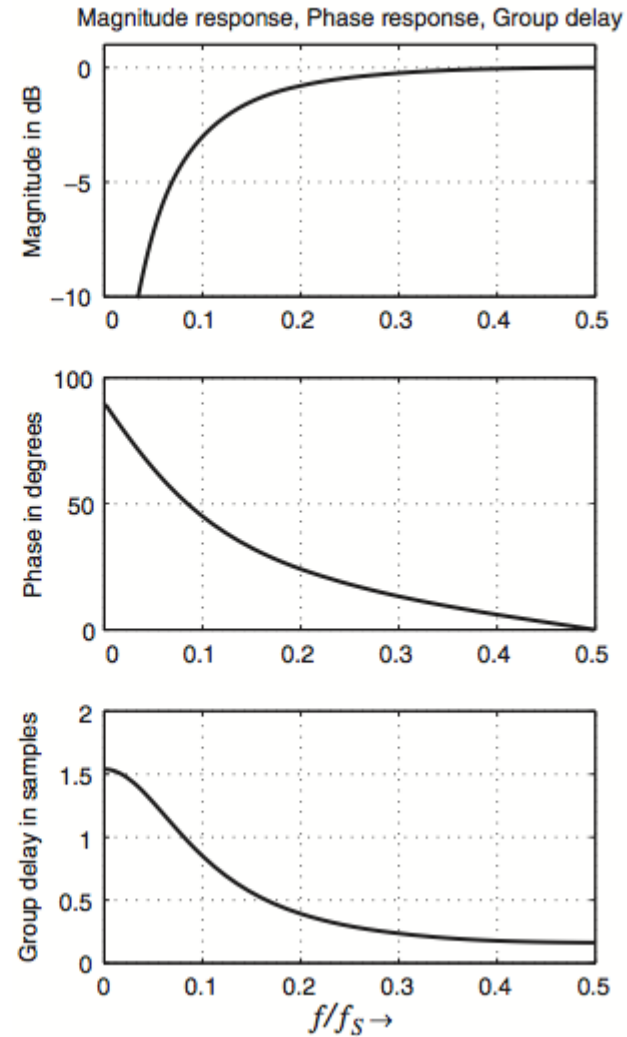
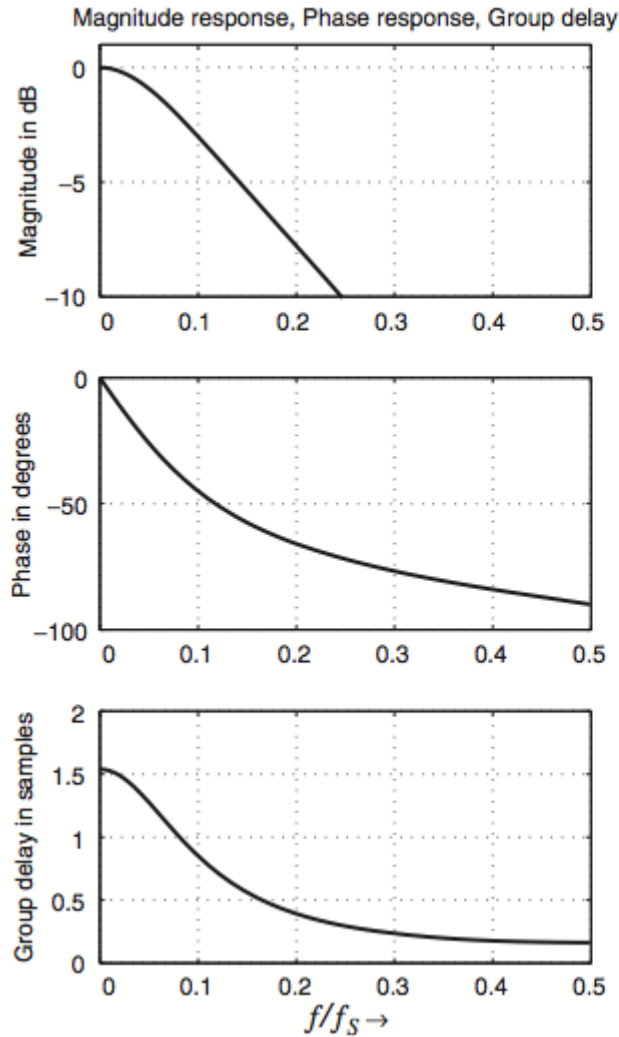
$$A(z) = \frac{z^{-1} + c}{1 + cz^{-1}}$$

$$c = \frac{\tan(\pi f_c/f_s) - 1}{\tan(\pi f_c/f_s) + 1}$$

Magnitude response, Phase response, Group delay



Allpass Low/Highpass



Universal Comb

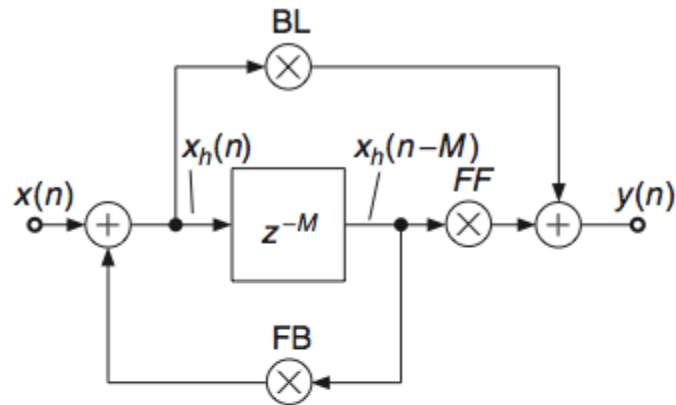


Figure 2.29 Universal comb filter.

Table 2.6 Parameters for universal comb filter.

	BL	FB	FF
FIR comb filter	1	0	g
IIR comb filter	c	g	0
Allpass	a	$-a$	1
Delay	0	0	1

Equalizers

- EQs shape the spectrum by enhancing (boost/cut) certain frequencies while others remain unaffected
- Usually built as a series of independently controlled 1st order Shelving and Peak filters

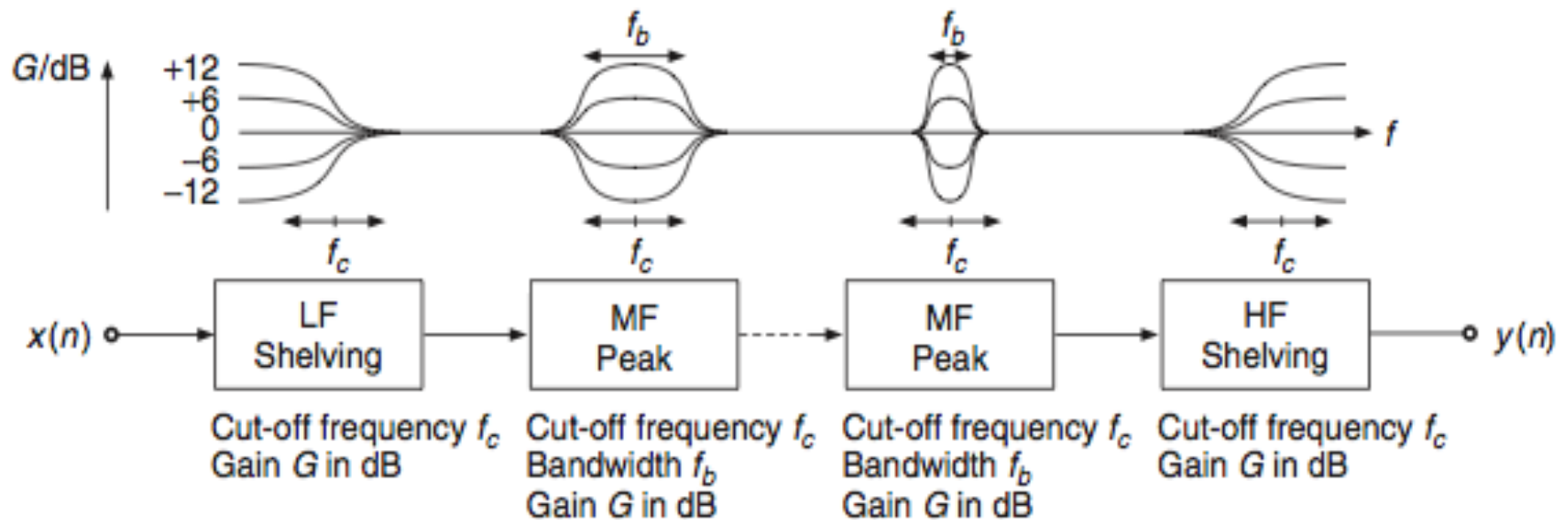


Figure 2.15 Series connection of shelving and peak filters.

Shelving Filter

- 1st-order design:

- Can be constructed based on a 1st-order allpass

$$H(z) = 1 + \frac{H_0}{2} [1 + \pm A(z)] \quad (\text{LF/HF} + / -)$$

- leads to the difference equation:

$$x_h(n) = x(n) - c_{B/C} x_h(n-1)$$

$$y_1(n) = c_{B/C} x_h(n) + x_h(n-1)$$

$$y(n) = \frac{H_0}{2} [x(n) \pm y_1(n)] + x(n).$$

- Gain (dB) can be adjusted through:

$$H_0 = V_0 - 1 \quad \text{with} \quad V_0 = 10^{G/20}.$$

- Frequency response slope limited to 6 dB / octave

Shelving coeffs

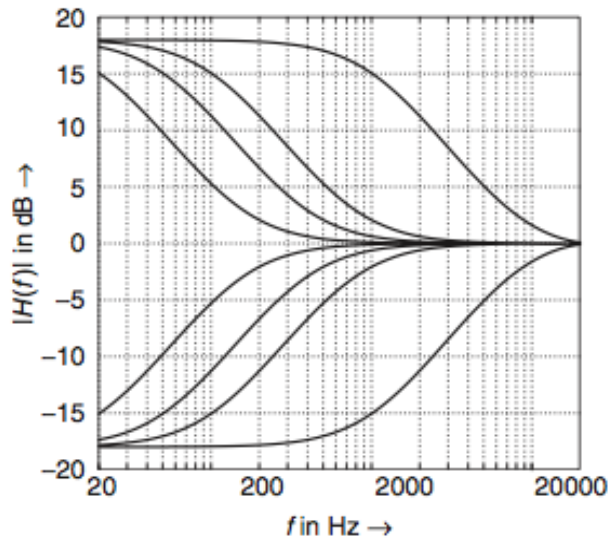
- Other parameters can be adjusted:

- c_B = boost frequency
- c_C = cut-off frequency

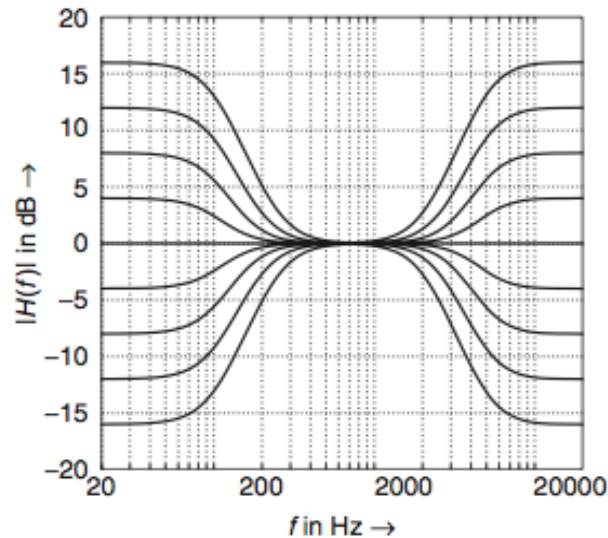
- 1st-order LF

- 1st-order HF

First-order shelving filters



Second-order shelving filters



$$\left[\begin{array}{l} c_B = \frac{\tan(\pi f_c/f_S) - 1}{\tan(\pi f_c/f_S) + 1}, \\ c_C = \frac{\tan(\pi f_c/f_S) - V_0}{\tan(\pi f_c/f_S) + V_0} \end{array} \right.$$

$$\left[\begin{array}{l} c_B = \frac{\tan(\pi f_c/f_S) - 1}{\tan(\pi f_c/f_S) + 1} \\ c_C = \frac{V_0 \tan(\pi f_c/f_S) - 1}{V_0 \tan(\pi f_c/f_S) + 1}. \end{array} \right.$$

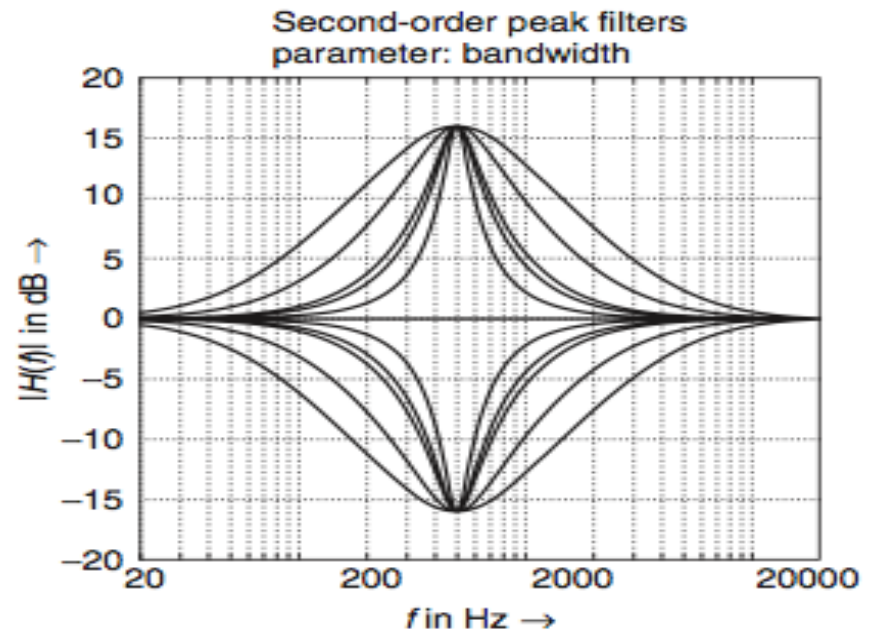
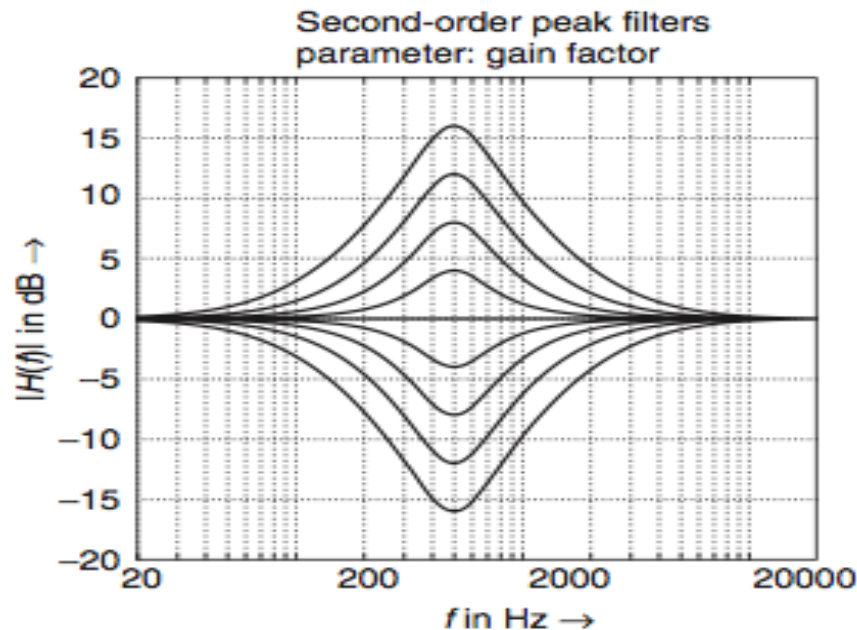
Figure 2.17 Frequency responses for first-order and second-order shelving filters.

Peak Filter

- 2nd order difference equation:

$$y(n) = \frac{H_0}{2} [x(n) - y_1(n)] + x(n).$$

- Offers almost independent control over the 3 musical parameters: f_c , f_b , G
- Q factor = f_b/f_c



Delay-based effects

- Example: Vibrato effect
 - Similar to Doppler, the idea that varying the distance (i.e. time delay) will affect the pitch
 - Periodical variation of the time delay produces a periodical pitch variation
 - Need a delay-line and LFO to drive delay time
 - typical params: LFO = 5-->14Hz / Width = 5-->10ms
 - Others effects include flanger, chorus, slapback, echo...

Table 2.8 Typical delay-based effects.

Delay range (ms) (Typ.)	Modulation (Typ.)	Effect name
0 ... 20	—	Resonator
0 ... 15	Sinusoidal	Flanging
10 ... 25	Random	Chorus
25 ... 50	—	Slapback
> 50	—	Echo