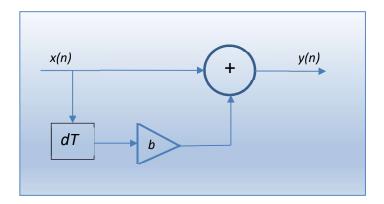
Audio Filters

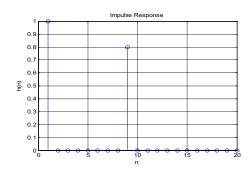
Basic echo filter (1st order filter FIR)

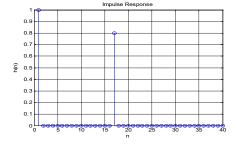
The filter below is a simple echo filter. It adds a copy of the input to itself dT seconds later, where d is the number of samples in the delay used.

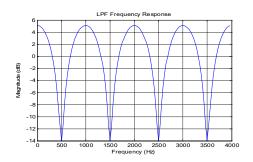


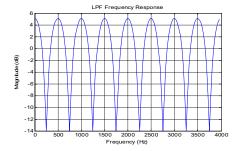
$$y(n) = x(n) + bx(n - d) \Rightarrow Y(z) = X(z) + bX(z)z^{-d}$$
$$\Rightarrow H(z) = \frac{Y(z)}{X(z)} = 1 + bz^{-d}$$

The impulse response and the frequency response of the echo filter are shown below for delays of 8 samples and 16 samples respectively.





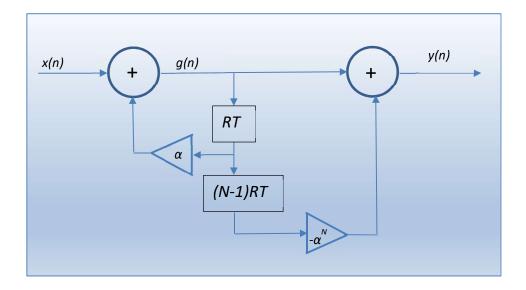




This filter called a *comb* filter because its frequency response resembles a comb.

$$\begin{split} &H(\theta) = 1 + be^{-j\theta d} = 1 + b\left\{\cos\left(\theta d\right) - j\sin\left(\theta d\right)\right\} \\ &\left|H(\theta)\right|^2 = \left\{1 + b\cos\left(\theta d\right)\right\}^2 + \left\{b\sin\left(\theta d\right)\right\}^2 = 1 + 2b\cos\left(\theta d\right) + b^2\left\{\cos^2\left(\theta d\right) + \sin^2\left(\theta d\right)\right\} \\ &\left|H(\theta)\right|^2 = 1 + 2b\cos\left(\theta d\right) + b^2 \\ &\left|H(\theta)\right|_{\max}^2 = 1 + 2b + b^2 = \left(1 + b\right)^2 \Rightarrow \left|H(\theta)\right|_{\max} = \left(1 + b\right) \\ &\left|H(\theta)\right|_{\min}^2 = 1 - 2b + b^2 = \left(1 - b\right)^2 \Rightarrow \left|H(\theta)\right|_{\min} = \left(1 - b\right) \\ &\left|H(\theta)\right|_{\max} @ \cos\left(\theta d\right) = 1 \Rightarrow \theta d = 2n\pi \Rightarrow \theta = \frac{2n\pi}{d} \\ &If d = 8, \theta = \frac{2n\pi}{8} = \frac{n\pi}{4} = n \times 1 \, kHz \, for \, f_s = 8 \, kHz \end{split}$$

Multiple echo filter



The multiple echo filter adds N-1 delays reducing in amplitude for each delay (α <1). The objective is to realise the following transfer function:

$$H(z) = \frac{Y(z)}{X(z)} = 1 + \alpha z^{-R} + \alpha^2 z^{-2R} + \alpha^3 z^{-3R} + \dots + \alpha^{N-1} z^{-(N-1)R}$$

We can however re-write this into a more compact form.

$$\begin{split} H(z) &= \frac{Y(z)}{X(z)} = 1 + \alpha z^{-R} + \alpha^2 z^{-2R} + \alpha^3 z^{-3R} + \dots + \alpha^{N-1} z^{-(N-1)R} \\ \alpha z^{-R} H(z) &= \alpha z^{-R} + \alpha^2 z^{-2R} + \alpha^3 z^{-3R} + \dots + \alpha^N z^{-NR} \\ H(z) &= \alpha z^{-R} H(z) = 1 + \alpha z^{-R} + \alpha^2 z^{-2R} + \alpha^3 z^{-3R} + \dots + \alpha^{N-1} z^{-(N-1)R} - \left\{ \alpha z^{-R} + \alpha^2 z^{-2R} + \alpha^3 z^{-3R} + \dots + \alpha^N z^{-NR} \right\} \\ H(z) \left(1 - \alpha z^{-R} \right) &= 1 - \alpha^N z^{-NR} \\ H(z) &= \frac{1 - \alpha^N z^{-NR}}{1 - \alpha z^{-R}} \end{split}$$

Now we need to verify that the above block diagram does indeed produce the compact transfer function derived above. Let the output from the 1^{st} summer be defined as q(n). Then:

$$G(z) = X(z) + \alpha G(z)z^{-R}$$

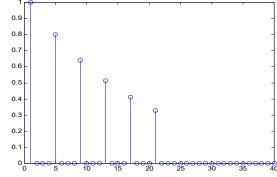
$$G(z) - \alpha G(z)z^{-R} = X(z)$$

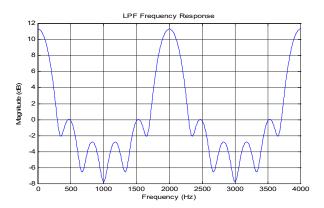
$$X(z) = G(z)\{1 - \alpha z^{-R}\}$$

$$Y(z) = G(z) - \alpha^{N}G(z)z^{-NR} = G(z)\{1 - \alpha^{N}z^{-NR}\}$$

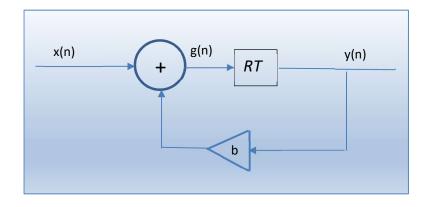
$$Y(z) = \frac{X(z)}{\{1 - \alpha z^{-R}\}}\{1 - \alpha^{N}z^{-NR}\}$$

$$H(z) = \frac{Y(z)}{X(z)} = \alpha z^{-R} + \alpha^{2}z^{-2R} + \alpha^{3}z^{-3R} + \dots + \alpha^{N}z^{-NR} = \frac{1 - \alpha^{N}z^{-NR}}{1 - \alpha z^{-R}}$$





Infinite echo filter



DSP3108 DSP Applications

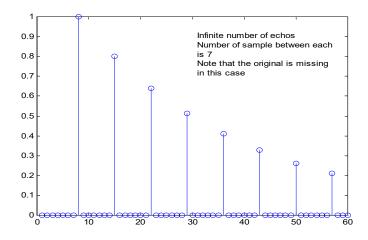
$$g(n) = x(n) + by(n)$$

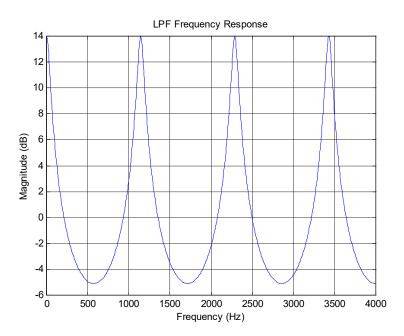
$$y(n) = g(n-R) = x(n-R) + by(n-R)$$

$$Y(z) = G(z)z^{-R} = X(z)z^{-R} + bY(z)z^{-R}$$

$$Y(z)(1-bz^{-R}) = X(z)z^{-R}$$

$$H(z) = \frac{Y(z)}{X(z)} = \frac{z^{-R}}{1-bz^{-R}}$$





It should be reasonably clear that this filter makes an infinite number of copies of itself scaled by 'b'.