Calculating Random Noise Attenuation Through a Lowpass Filter

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March 21, 2008

Given the following discrete-time lowpass filter:

$$y(t_k) = (1 - a)y(t_{k-1}) + au(t_k)$$
(1)

If this filter stems from discretizing the continuous-time filter

$$H(s) = \frac{1}{T_f s + 1} \tag{2}$$

with the Euler Backward Difference method with time-step h [s], then

$$a = \frac{h}{T_f + h} \tag{3}$$

Inversely,

$$T_f = h \frac{1-a}{a} \tag{4}$$

We will now analyze the discrete-time lowpass filter (1). We assume that the filter input, u, is a random signal with standard deviation σ_u and hence variance

$$Var(u) = \sigma_u^2 \tag{5}$$

We will now calculate the stationary variance of the filter output, y. This variance will be a function of the filter parameter a and the sampling interval h. This function will then be solved for a so that we get a formula a. In the following the following simplifying notation will be used:

$$x_k = x(t_k) \tag{6}$$

The variance is actually the expected value of the square of the signal value (we assume here that the mean value of the signal considered is zero):

$$Var [y_k] = \sigma_y^2 = E(y_k^2)$$
(7)

From (1) the output variance is

$$E(y_k^2) = E\{[(1-a)y_{k-1} + au_k][(1-a)y_{k-1} + au_k]\}$$
(8)

$$= E[(1-a)^2 y_{k-1}^2] + E[2(1-a)y_{k-1}au_k] + E[a^2 u_k^2]$$
(9)

$$= (1-a)^2 \underbrace{E[y_{k-1}^2]}_{=E[y_k^2]} + 2(1-a)\underbrace{a\underbrace{E[y_{k-1}u_k]}_{=0}}_{=0} + a^2 \underbrace{E[u_k^2]}_{=\sigma_u^2}$$
(10)

$$= (1-a)^2 E[y_k^2] + a^2 \sigma_u^2$$
(11)

In (10) $E[y_{k-1}u_k] = 0$ because y_{k-1} and u_k are uncorrelated (independent). In (10) $E[y_{k-1}^2] = E[y_k^2]$ because it is assumed that the signals are stationary. Solving (11) for $E(y_k^2)$ gives

$$E(y_k^2) = \sigma_y^2 = \frac{a^2}{1 - (1 - a^2)} \sigma_u^2 = \frac{a^2}{(2a - a^2)} \sigma_u^2 = \frac{a}{2 - a} \sigma_u^2$$
 (12)

Thus, the ratio between the output standard deviation and the input standard deviation is

$$\frac{\sigma_y}{\sigma_u} = \sqrt{\frac{a}{2-a}} \stackrel{\text{def}}{=} K_\sigma \tag{13}$$

where K_{σ} is the ratio between the output and input standard deviations. So, once a is given, K_{σ} can be calculated.

You can also go the opposite way: Solving (13) for a gives

$$a = \frac{2K_{\sigma}^2}{1 + K_{\sigma}^2} \tag{14}$$

So, once you have specified K_{σ} , you know what filter parameter to use. If you want the filter time-constant T_f in stead of a, you can use (4).