

# Solution to ECE Test 13 S09

We have studied the following seven digital filter design techniques:

Impulse Invariant (II), Step Invariant (SI), Finite Difference (FD), Direct Substitution (DS), Matched  $z$  (MZ), Bilinear (BL) and Windowing an Ideal Impulse Response (WI).

I. Answer the following questions using the two-letter acronyms above.

1. Which of the techniques produce an IIR digital filter ?

All except WI.

2. Which of the techniques use a transformation from  $s$  to  $z$  to convert an analog filter to a digital filter without directly using any time-domain functions?

FD, DS, MZ, BL

3. Which techniques guarantee that an analog lowpass filter will be transformed into a digital filter whose response at  $\Omega = \pm\pi$  is zero?

BL

II.

1. An analog filter has a transfer function  $H(s) = \frac{s^2 - 2}{s^2 + 4s + 3}$ . A digital filter is designed by the direct substitution method with a sampling rate  $f_s = 5$  samples/second. List the numerical locations of all its poles and zeros in the  $z$  plane.

In the  $s$  plane the zeros are at  $\pm\sqrt{2}$  and the poles are at  $-1, -3$ .

In the  $z$  plane the zeros are at  $e^{\pm\sqrt{2}/5} = 1.3269$  and  $0.7536$ .

In the  $z$  plane the poles are at  $e^{-1/5} = 0.8187$  and  $e^{-3/5} = 0.5488$

2. An analog filter has a transfer function  $H(s) = 1/s$ . A digital filter is designed by the bilinear method with a sampling rate  $f_s = 30$  samples/second. List the numerical locations of all its poles and zeros in the  $z$  plane.

$$H(z) = \left[ 1/s \right]_{s \rightarrow \frac{2}{T_s} \frac{z-1}{z+1}} = \frac{T_s}{2} \frac{z+1}{z-1} = \frac{1}{60} \frac{z+1}{z-1}$$

In the  $z$  plane the zero is at  $z = -1$ .

In the  $z$  plane the pole is at  $z = 1$ .

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BL

II.

1. An analog filter has a transfer function  $H(s) = \frac{s^2 + 2}{s^2 + 6s + 5}$ . A digital filter is designed by the direct substitution method with a sampling rate  $f_s = 5$  samples/second. List the numerical locations of all its poles and zeros in the  $z$  plane.

In the  $s$  plane the zeros are at  $\pm j\sqrt{2}$  and the poles are at  $-1, -5$ .

In the  $z$  plane the zeros are at  $e^{\pm j\sqrt{2}/5} = e^{\pm j0.2828} = 0.9603 \pm j0.2791$ .

In the  $z$  plane the poles are at  $e^{-1/5} = 0.8187$  and  $e^{-5/5} = 0.3678$

2. An analog filter has a transfer function  $H(s) = 1/s$ . A digital filter is designed by the bilinear method with a sampling rate  $f_s = 30$  samples/second. List the numerical locations of all its poles and zeros in the  $z$  plane.

$$H(z) = \left[ 1/s \right]_{s \rightarrow \frac{2}{T_s} \frac{z-1}{z+1}} = \frac{T_s}{2} \frac{z+1}{z-1} = \frac{1}{60} \frac{z+1}{z-1}$$

In the  $z$  plane the zero is at  $z = -1$ .

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BL

II.

1. An analog filter has a transfer function  $H(s) = \frac{s^2 + 1}{s^2 + 8s + 7}$ . A digital filter is designed by the direct substitution method with a sampling rate  $f_s = 5$  samples/second. List the numerical locations of all its poles and zeros in the  $z$  plane.

In the  $s$  plane the zeros are at  $\pm j$  and the poles are at  $-1, -7$ .

In the  $z$  plane the zeros are at  $e^{\pm j/5} = 0.980 \pm j0.199$ .

In the  $z$  plane the poles are at  $e^{-1/5} = 0.8187$  and  $e^{-7/5} = 0.2466$

2. An analog filter has a transfer function  $H(s) = 1/s$ . A digital filter is designed by the bilinear method with a sampling rate  $f_s = 30$  samples/second. List the numerical locations of all its poles and zeros in the  $z$  plane.

$$H(z) = \left[ 1/s \right]_{s \rightarrow \frac{2z-1}{T_s z+1}} = \frac{T_s z+1}{2z-1} = \frac{1}{60} \frac{z+1}{z-1}$$

In the  $z$  plane the zero is at  $z = -1$ .

In the  $z$  plane the pole is at  $z = 1$ .