

Matched Z Transform $\rightarrow$

In this analog

filter is transformed into the digital filter by mapping poles and zeros of $H(s)$ directly into the poles and zeros in z plane. Consider that the transfer fn of analog filter is given by

$$H(s) = \frac{\sum_{k=1}^M (s - z_k)}{\sum_{k=1}^M (s - p_k)} \quad (1)$$

Here z_k represents the zeros of system and p_k poles of system

In this method the poles obtained from matched z transform are identical to poles obtained with impulse invariance method. But the zeros positions are different than impulse invariance method.

The transfer fn of digital filter can be given by

$$H(z) = \frac{\sum_{k=1}^M (1 - e^{z_k T_s} \cdot z^{-1})}{\sum_{k=1}^M (1 - e^{p_k T_s} \cdot z^{-1})}$$

Here T_s is the sampling time. In this method, to maintain the freq response characteristics of analog filter normal, the sampling interval (T_s) is kept very low in order to avoid aliasing.

Butterworth Filter $\Rightarrow$

design eq'ns and design steps

$A_p =$ Attenuation in pass band

$A_s =$ " " stop "

$\Omega_p =$ Pass band edge freq

$\Omega_c =$ Cut off freq

$\Omega_s =$ stop band edge freq

Step 1 From the given specifications of digital filter obtain equivalent analog filter

(a) for impulse invariance method

$$\Omega = \frac{\omega}{T_s}$$

(b) for BCT $\Rightarrow \Omega = \frac{2}{T_s} \tan \frac{\omega}{2}$

$\Omega =$ freq of analog filter

$\omega =$ freq of digital filter

$T_s =$ sampling time

Step 2

Calculate the order N of filter using the equation

$$N = \frac{1}{2} \times \frac{\log \left[\frac{\left(\frac{1}{A_s^2} - 1 \right)}{\left(\frac{1}{A_p^2} - 1 \right)} \right]}{\log \left(\frac{\omega_s}{\omega_p} \right)}$$

or $N = \frac{\log \left[\frac{(1 - S_s^2) - 1}{2 \log(\omega_s/\omega_c)} \right]}{\log(\omega_s/\omega_p)}$ Here S_s = Attenuation in stop band

If the specifications are given in decibels (dB) then use the eq'n

$$N = \frac{1}{2} \times \frac{\log \left[\frac{10^{0.1 A_s(\text{dB})} - 1}{10^{0.1 A_p(\text{dB})} - 1} \right]}{\log \left(\frac{\omega_s}{\omega_p} \right)}$$

Step 3

Calculations of Cut off freq (ω_c)
The cut-off freq ω_c of analog filter is calculated as:-

(a) for IIR $\Rightarrow \omega_c = \frac{\omega_c}{T_s}$

(b) for BLT $\omega_c = \frac{2}{T_s} \tan \frac{\omega_c}{2}$

when ω_c is not given then use the eq'n

(i) $\omega_c = \frac{\omega_p}{\left(\frac{1}{A_p^2} - 1 \right)^{\frac{1}{2N}}}$ and if specifications are in dB then

(ii)

$$\rho_c = \frac{\rho_p}{10^{0.1AP} - 1}$$

Step 4

Calculate the poles using

$$P_k = \pm \rho_c e^{j(N+2k+1)\pi/2N} \quad k=0, 1, 2, \dots, N-1$$

If the poles are complex conjugate then organize the poles (P_k) as complex conjugate pairs that means

$$S_1 \text{ and } S_1^* \quad S_2 \text{ and } S_2^*$$

Step 5

Calculate the system transfer fn of analog filter using

$$H(s) = \frac{\rho_c^N}{(s-P_1)(s-P_2)}$$

and if poles are complex conjugate then

$$H(s) = \frac{\rho_c^N}{(s-S_1)(s-S_1^*)(s-S_2)(s-S_2^*)}$$

Step 6

Design the digital filter using ILM or BCT

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A digital filter has freq specification as

$$\text{Passband freq} = \omega_p = 0.2\pi$$

$$\text{Stopband freq} = \omega_s = 0.3\pi$$

What are the corresponding specification

- ① ILM is used for designing
- ② BCT is used for designing

Soln Assume sampling $T_s = 1$

For IIM $\omega_p = \frac{\omega_p}{T_s} \quad \therefore \omega_p = 0.2\pi$
 $= 0.63 \text{ rad/sec}$

$$\omega_s = \frac{\omega_s}{T_s} = \omega_s = \frac{0.3\pi}{T_s} = 0.3\pi$$

$$\omega_s = 0.94 \text{ rad/sec}$$

For BCT $\omega_p = \frac{2}{T_s} \tan\left(\frac{\omega_p}{2}\right)$

$$\omega_p = 2 \tan\left(\frac{0.2\pi}{2}\right) = 2 \tan\left(\frac{0.2 \times 180}{2}\right)$$

$$\omega_p = 0.65 \text{ rad/sec}$$

$$\omega_s = \frac{2}{T_s} \tan\left(\frac{\omega_s}{2}\right)$$

$$\omega_s = 2 \tan\left(\frac{0.3\pi}{2}\right) = 2 \tan\left(\frac{0.3 \times 180}{2}\right)$$

$$\omega_s = 1.019 \text{ rad/sec}$$

Using bilinear transformation design a butterworth filter which satisfies the following conditions

$$0.8 \leq |H(e^{j\omega})| \leq 1$$

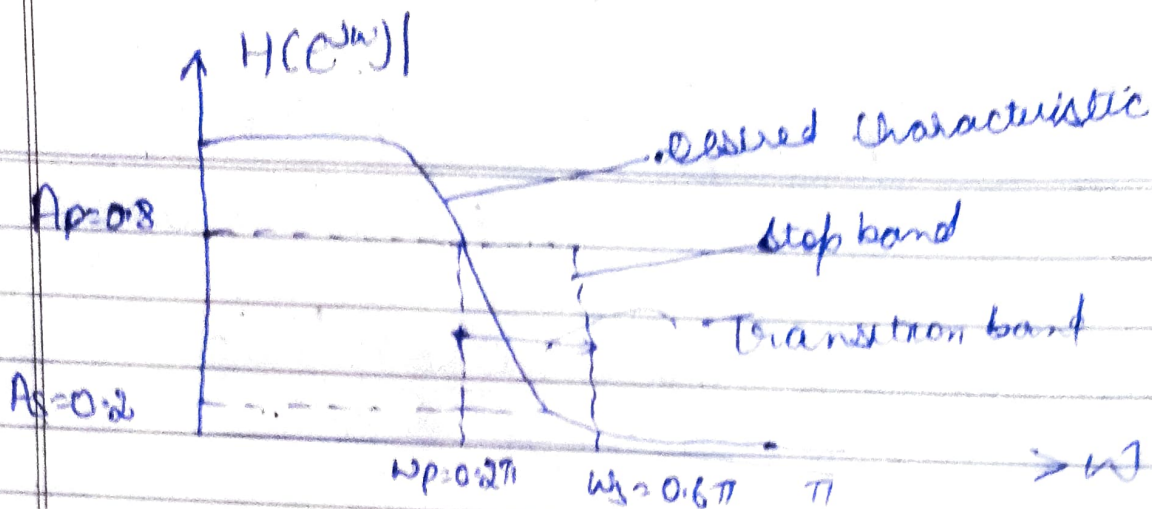
$$0 \leq \omega \leq 0.2\pi$$

$$|H(e^{j\omega})| \leq 0.2$$

$$0.6\pi \leq \omega \leq \pi$$

Soln

$H(e^{j\omega})$ is same as $|H(\omega)|$; so.



Given specifications of digital filter are

$$A_p = 0.8, \quad \omega_p = 0.2\pi, \quad A_s = 0.2, \quad \omega_s = 0.6\pi$$

Part A

Step 1 Calculating specifications of analog filter.

We will obtain the values of Ω_p and Ω_s .
We have to use BLT

$$\Omega = \frac{2}{T_s} \tan \frac{\omega}{2}$$

$$\Omega_p = \frac{2}{T_s} \tan \frac{\omega_p}{2}$$

The value of sampling time T_s is not given so assume $T_s = 1$

$$\Omega_p = 2 \tan \frac{0.2\pi}{2}$$

$$\therefore \Omega_p = 0.65$$

$$\text{Similarly } \Omega_s = \frac{2}{T_s} \tan \frac{\omega_s}{2} = 2 \tan \frac{0.6\pi}{2}$$

$$\Omega_s = 2.75$$

Thus specifications of analog filter are
 $A_p = 0.8$ $A_p = 0.65$ $A_s = 0.2$ $\omega_s = 2.75$

Step 2 Calculation of order of filter N
It is given by

$$N = \frac{1}{2} \times \frac{\log \left[\left(\frac{1}{A_s^2} - 1 \right) / \left(\frac{1}{A_p^2} - 1 \right) \right]}{\log \left(\frac{\omega_s}{\omega_p} \right)}$$

$$N = \frac{1}{2} \times \frac{\log \left[\left(\frac{1}{(0.2)^2} - 1 \right) / \left(\frac{1}{(0.8)^2} - 1 \right) \right]}{\log \left(\frac{2.75}{0.65} \right)}$$

$$= \frac{1}{2} \times \frac{\log \left[\frac{24}{0.5625} \right]}{\log \left(\frac{2.75}{0.65} \right)} = 1.3$$

approximately we can take $N=2$

Step 3 Calculating of cut-off freq ω_c

ω_c is not given so use the eqn

$$\omega_c = \frac{\omega_p}{\left(\frac{1}{A_p^2} - 1 \right)^{1/2N}} = \frac{0.65}{\left[\frac{1}{(0.8)^2} - 1 \right]^{1/4}} = 0.75$$

Step 4

Calculations of the poles of $H(s)$

$$P_k = \pm \rho_k e^{j(N+2k+1)\pi/2\pi}$$

$$K=0, 1, \dots, N-1$$

$$P_k = \pm 0.75 e^{j(2+2K+1)\pi/4}$$

$$P_k = \pm 0.75 e^{j(3+2K)\pi/4}$$

Here range of K is from 0 to $N-1$ and $N=2$, thus the value of K is $K=0$ and $K=1$

For $K=0$

$$P_0 = \pm 0.75 e^{j(3)\pi/4}$$

$$= \pm 0.75 \left[\cos\left(\frac{3\pi}{4}\right) + j \sin\left(\frac{3\pi}{4}\right) \right]$$

$$P_0 = \pm 0.75 [-0.707 + j0.707]$$

$$P_0 = \pm [-0.53 + j0.53]$$

$$P_0 = -0.53 + j0.53 \quad \text{and} \quad 0.53 - j0.53$$

For $K=1$

$$P_1 = \pm 0.75 e^{j(3+2)\pi/4} = \pm 0.75 e^{j5\pi/4}$$

$$P_1 = \pm 0.75 \left[\cos\frac{5\pi}{4} + j \sin\frac{5\pi}{4} \right]$$

$$P_1 = \pm 0.75 [-0.707 - j0.707]$$

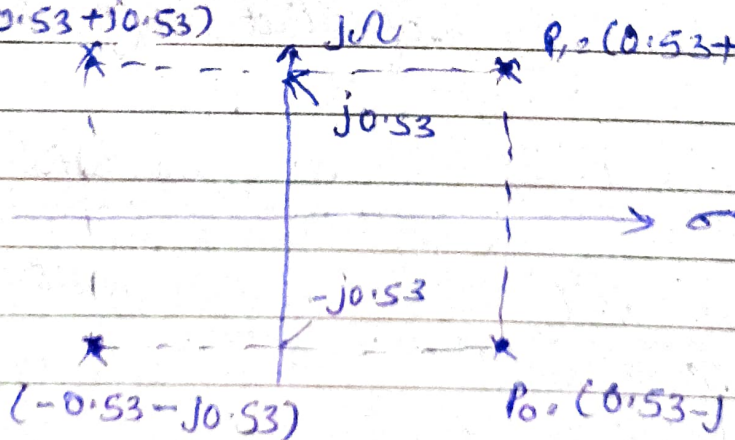
$$P_1 = -0.53 - j0.53 \quad \text{and} \quad 0.53 + j0.53$$

$$P_0 = (-0.53 + j0.53)$$

$$P_1 = (0.53 + j0.53)$$

$$P_1 = (-0.53 - j0.53)$$

$$P_0 = (0.53 - j0.53)$$



$$s_1 = -0.53 + j0.53$$

$$s_1^* = -0.53 - j0.53$$

Step 5 Calculation of $H(s)$

The transfer fn of analog filter is given by

$$H(s) = \frac{0.6N}{(s-s_1)(s-s_1^*)}$$

$$H(s) = \frac{(0.75)^2}{(s+0.53-j0.53)(s+0.53+j0.53)}$$

$$H(s) = \frac{(0.75)^2}{s^2 + 0.53s + j0.53s + 0.53s + 0.28 + j0.28 - j0.53s - j0.28 + 0.28}$$

$$H(s) = \frac{(0.75)^2}{s^2 + 1.06s + 0.56}$$

$$H(s) = \frac{0.56}{s^2 + 1.06s + 0.56}$$

Part B we have to use BLT. Then $H(z)$ is obtained by putting $s = \frac{z-1}{T_s}$ in the eq'n $H(s)$.

$$H(z) = \frac{0.56}{4\left(\frac{z-1}{z+1}\right)^2 + (2 \times 1.06)\left(\frac{z-1}{z+1}\right) + 0.56}$$

$$H(z) = \frac{0.56}{\frac{4(z-1)^2}{(z+1)^2} + \frac{2.12z - 2.12}{z+1} + 0.56}$$

$$H(Z) = \frac{0.56(Z+1)^2}{4(Z^2 - 2Z + 1) + (2.12Z - 2.212)(Z+1) + 0.56(Z+1)^2}$$

$$H(Z) = \frac{0.56(Z+1)^2}{4Z^2 - 8Z + 4 + 2.12Z^2 + 2.12Z - 2.12Z - 2.12 + 0.56Z^2 + 1.12Z + 0.56}$$

$$H(Z) = \frac{0.56(Z+1)^2}{6.68Z^2 - 6.88Z + 2.44}$$

Analog and digital Transformation in freq Domain $\Rightarrow$

Freq transformation in Analog Domain

we have normalized LPT having cut-off freq Ω_c

Low pass to low Pass $\Rightarrow$

$$s = \frac{\Omega_c}{\Omega_{LP}} S$$

Low Pass to high Pass $\Rightarrow$

HPP with cutoff freq Ω_{HP}

$$s = \frac{\Omega_c \Omega_{HP}}{S}$$

3. Low Pass to Band Pass

$$S \Rightarrow \Omega_c \frac{S^2 + \Omega_c \Omega_u}{S(\Omega_u - \Omega_c)}$$

4. Lowpass to bandstop or notch or band reject

$$S \Rightarrow \Omega_c \frac{S(\Omega_u - \Omega_c)}{S^2 + \Omega_c \Omega_u}$$

Frequency Transformation in Digital Domain $\rightarrow$

1. Low Pass to Low Pass $\Rightarrow$

$$Z^{-1} \Rightarrow \frac{Z^{-1}d}{1 - dZ^{-1}}$$

$$\text{Here } d = \frac{\sin\left(\frac{\omega'_p - \omega_p}{2}\right)}{\sin\left(\frac{\omega'_p + \omega_p}{2}\right)}$$

ω'_p = Pass band freq. of LPP

ω_p = " " " " new filter

2. Low pass to high Pass

$$Z^{-1} \rightarrow \frac{-Z^{-1} + d}{1 + dZ^{-1}}$$

$$\text{Here } d = \frac{-\cos\left(\frac{\omega'_p + \omega_p}{2}\right)}{\cos\left(\frac{\omega'_p - \omega_p}{2}\right)}$$

$$\cos\left(\frac{\omega'_p - \omega_p}{2}\right)$$

ω'_p $\rightarrow$ Pass band freq. of low Pass filter

ω_p = " " " " High " "

3 Low pass to band pass $\rightarrow$

$$Z^{-1} \rightarrow -Z^{-2} - \frac{2\alpha K}{K+1} Z^{-1} + \frac{K-1}{K+1}$$

$$\frac{\frac{K-1}{K+1} Z^{-2} - \frac{2\alpha K}{K+1} Z^{-1} + 1}{1}$$

$$\alpha = \frac{\cos\left(\frac{\omega_2 + \omega_1}{2}\right)}{\cos\left(\frac{\omega_2 - \omega_1}{2}\right)}$$

$$\text{And } K = \cot\left(\frac{\omega_2 - \omega_1}{2}\right) \tan \frac{\omega_p}{2}$$

4 Low Pass to band reject $\rightarrow$

$$Z^{-1} \rightarrow Z^2 - \frac{2\alpha}{K+1} Z^{-1} + \frac{1-K}{1+K}$$

$$\frac{\frac{1-K}{1+K} Z^{-2} - \frac{2\alpha}{1+K} Z^{-1} + 1}{1}$$

$$\text{Here } \alpha = \frac{\cos\left(\frac{\omega_2 + \omega_1}{2}\right)}{\cos\left(\frac{\omega_2 - \omega_1}{2}\right)}$$

$$K = \tan\left(\frac{\omega_2 - \omega_1}{2}\right) \tan \frac{\omega_p}{2}$$

Design the digital high pass filter for cut off freq of 30Hz and sampling freq of 150Hz using BLT

Soln

Part A Calculations of specifications for digital filter

$$f_{HP} = \frac{F_{HP}}{F_s}$$

$$F_{HP} = 30 \text{ Hz}$$

$$F_s = 150 \text{ Hz}$$

$$f_{HP} = \frac{30}{150} = 0.2 \text{ cycles/Samples}$$

$$\omega_{HP} = 2\pi f_{HP} = 2\pi \times 0.2 = 0.4\pi$$

The order of filter is not given, so we can assume order of $N=1$

Part B

Calculation of specifications for analog filter

Step 1

Calculation of ω_{HP}

$$\omega = \frac{\omega_p}{T_s} \tan \frac{\omega}{2}$$

$$\omega_{HP} = \frac{2}{\frac{1}{150}} \tan \frac{\omega_{HP}}{2} = 300 \tan \frac{0.4\pi}{2}$$

$$\omega_{HP} = 217.96 \text{ rad/sec}$$

we assumed $N=1$

fn of prototype L.P.F

$$H(s) = \frac{1}{s-1} = \frac{1}{s+1}$$

Part C

use freq transformation

$$s \rightarrow \frac{\omega_c \omega_{HP}}{s}$$

$$\omega_c = 1$$

$$s = \frac{\omega_{HP}}{s} \Rightarrow s = \frac{217.96}{s}$$

Putting this value in eqⁿ ① we get

$$H(s) = \frac{1}{\frac{217.96}{s} + 1} = \frac{s}{s + 217.96} \quad \text{--- (2)}$$

Part D Calculation of $H(z)$ for digital filter using BLT

$$s = \frac{2}{T_s} \left[\frac{z-1}{z+1} \right] \text{ in eqⁿ (2)}$$

$$H(z) = \frac{2 \times 150 \left(\frac{z-1}{z+1} \right)}{2 \times 150 \left(\frac{z-1}{z+1} \right) + 217.96} = \frac{300 \left(\frac{z-1}{z+1} \right)}{300 \left(\frac{z-1}{z+1} \right) + 217.96}$$

$$H(z) = \frac{\left(\frac{z-1}{z+1} \right)}{\left(\frac{z-1}{z+1} \right) + 0.73}$$

$$H(z) = \frac{z-1}{\{z-1\} + 0.73z + 0.73}$$

$$H(z) = \frac{z-1}{1.73z - 0.27}$$

Comparison b/w IIR and FIR filter

Char	IIR	FIR
Number of necessary multiplications	Least	Most
Sensitivity of filter coefficient quantization	Can be high for direct form	Very low
Probability of overflow error	Can be high for direct form	Very low
Stability	Depended upon system design	Guaranteed
Linear Phase	No	Guaranteed
Can simulate prototype analog filter	Yes	No
Required memory	Least	Most
How filter control complexity	Moderate	Simple
Availability of design software	Good	Very good