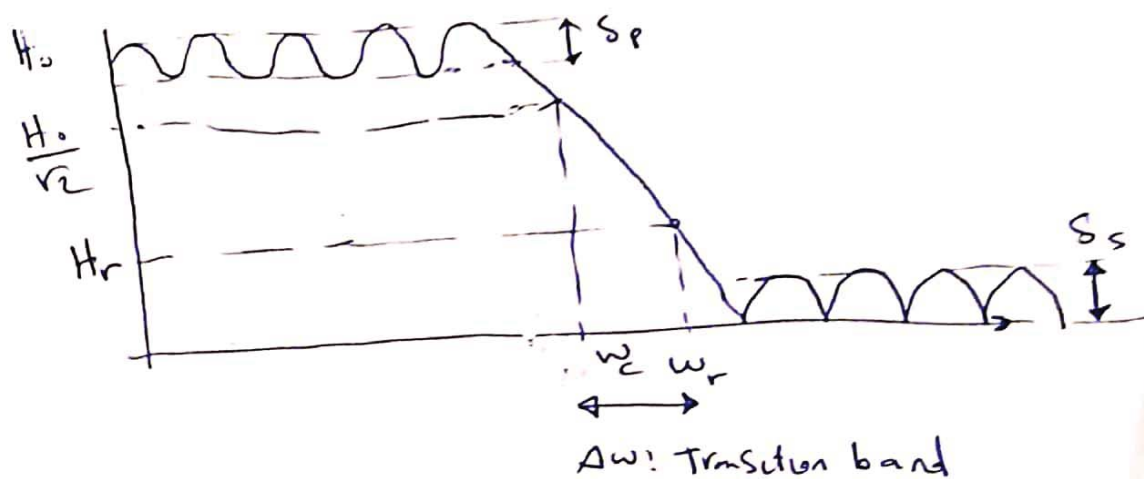


Design of Digital Filters

Given:

Desired Frequency domain characteristics of the Filter. This includes

- [1] Type of Filter IIR or FIR?
- [2] LPF or HPF or BPF or BSF?
- [3] Type of Approximation
[Butterworth, chebyshev I, chebyshev II, Elliptic]
- [4] Specifications of Pass Band & Stop Band



S_p : passband ripples

S_s : Stop band ripples

ω_c : cut-off frequency

ω_r : rejection frequency

H_r : magnitude at ω_r

ΔW : Transition B.W

Required

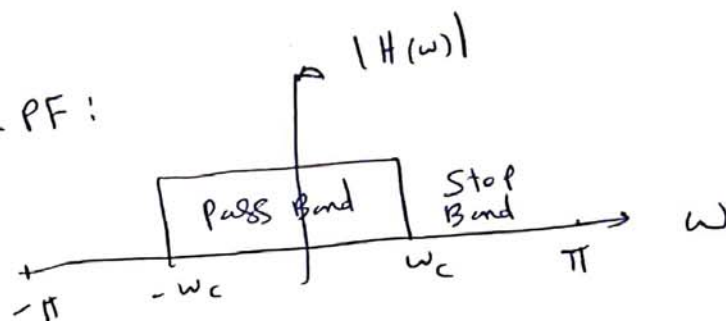
Find the coefficient (a 's & b 's) of the Causal Filter that meet the desired frequency domain specification.

This includes

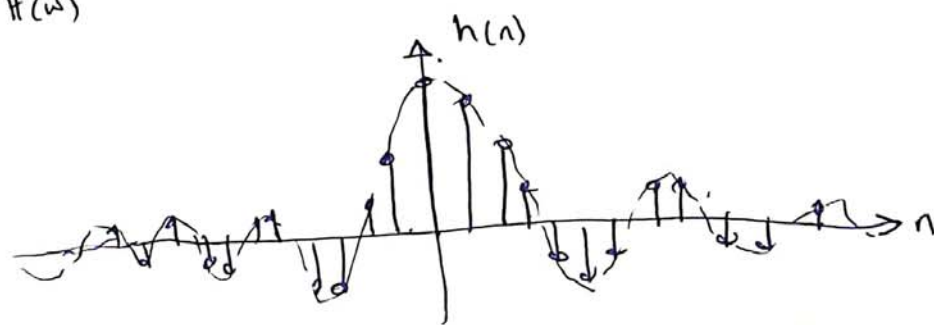
- 1 order (N)
- 2 $H(z)$ of the filter
- 3 a 's & b 's coefficients
- 4 Realization of filter.

1 @ Why Ideal Filters can not work?

For Ideal LPF:



rectangular in Frequency Domain $H(\omega)$ \Rightarrow Sinc in time Domain $h(n)$



Non causal why?

$$h(n) \neq 0, n < 0$$

The Filter depends on future input values

Therefore, Ideal Filters can not be implemented

Q: Compare FIR Filter with IIR Filter?

FIR	IIR
<p>[1] FIR can fulfill <u>Linear phase</u> requirements in passband</p>	<p>[1] It is very hard to design stable IIR Filter with linear phase.</p>
<p>[2] D.E: $y(n) = a_0 x(n) + a_1 x(n-1) + a_2 x(n-2) - \dots$</p> <p>No Feed back, output depends only on input</p> <p>[more simple]</p>	<p>[2] D.E: $y(n) + b_1 y(n-1) + b_2 y(n-2) = a_0 x(n) + a_1 x(n-1) + \dots$</p> <p>Feed back, output depends on input & past output</p>
<p>[3] Always Stable</p>	<p>[3] May be unstable</p>
<p>[4] Unrelated to Continuous time Filtering</p>	<p>[4] Mainly derived from Analog prototypes</p>
<p>[5] High order to accomplish different filtering tasks</p> <p>[disadvantage]</p>	<p>[5] Low order to achieve the same objective.</p> <p>typically less than $\frac{1}{10}$ of FIR Filter [Advantage]</p> <p>$N _{IIR} < \frac{1}{10} N _{FIR}$</p> <p>[To achieve the same requirements]</p>

Filter Design Techniques

FIR Design

Windowing Method

IIR Design

• Continuous to Discrete transformation

[1] Bilinear Transformation

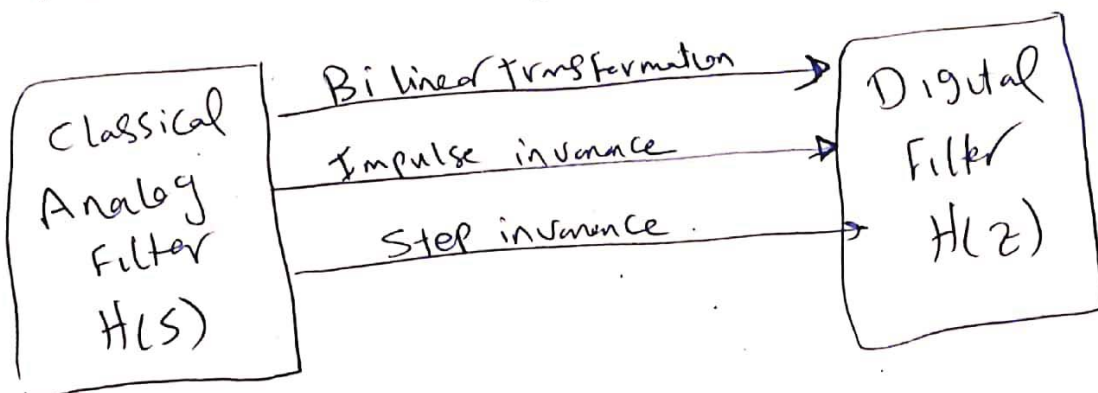
[2] Impulse Invariance

[3] Step Invariance

Design of IIR Digital Filters

- In IIR Design, we start the design in the analog plane, then make transformation from Analog to digital plane.

- There are many transfer methods but we will study 3 methods [Bilinear Transformation, Impulse invariance, Step invariance]



In each type, Filter order is a major concern.

As the order of filter $N \uparrow \uparrow$

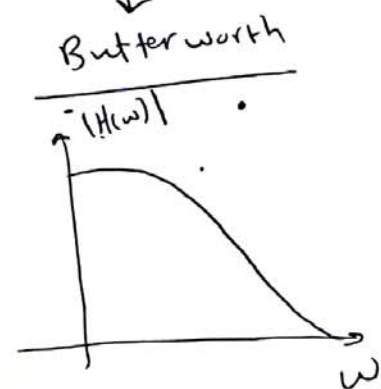


Achieve specifications better

but also more complicated & prone to rounding errors.

The First step in IIR design is to get the design of the analog normalized LPF? why?

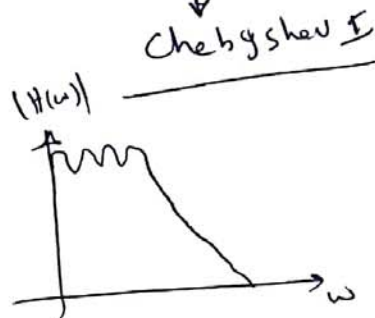
Analog Filters have a variety of well-established methods for design



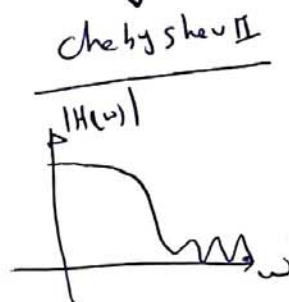
Δw largest

Flat in both pass & stop Bands

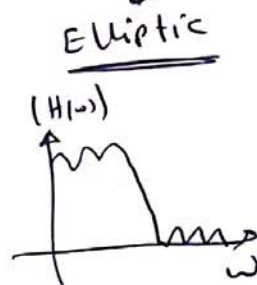
[We will study this only]



Δw medium ripples in pass band



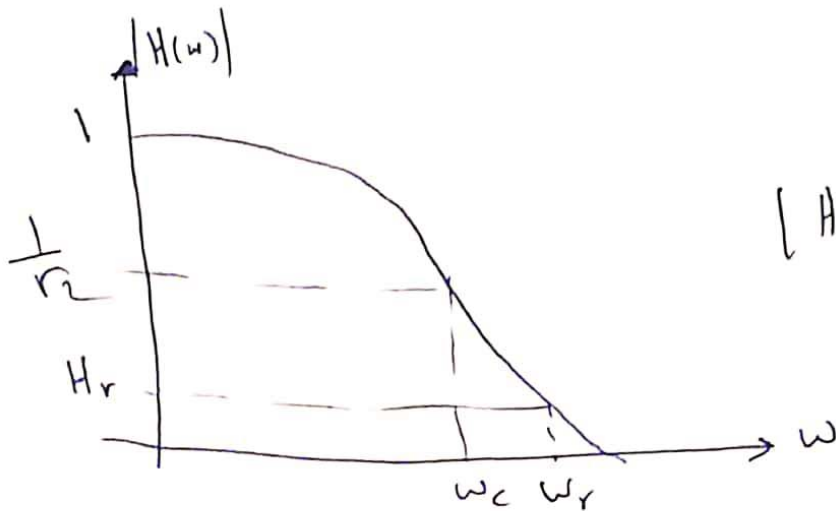
Δw medium ripples in stop band



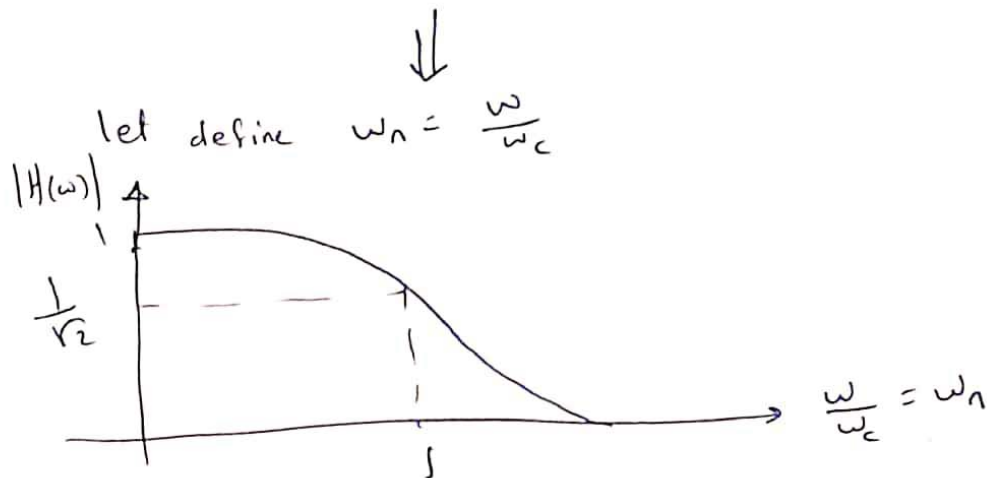
Δw smallest ripples in both pass & stop Bands

out of scope of this course.

Butterworth Analog Design



$$|H(\omega)|^2 = \frac{1}{1 + \left(\frac{\omega}{\omega_c}\right)^{2N}}$$



$$|H(\omega)|^2 = \frac{1}{1 + \omega_n^{2N}}, \quad N: \text{order of the filter.}$$

(Note) $|H(\omega)|^2 = H(j\omega) \cdot H(-j\omega) = \frac{1}{1 + (\omega_n^2)^N}$

↓

$$H(s) \cdot H(s) = \frac{1}{1 + (-s^2)^N} \quad \text{because } s = j\omega, \quad s^2 = (j\omega)^2 = -\omega^2$$

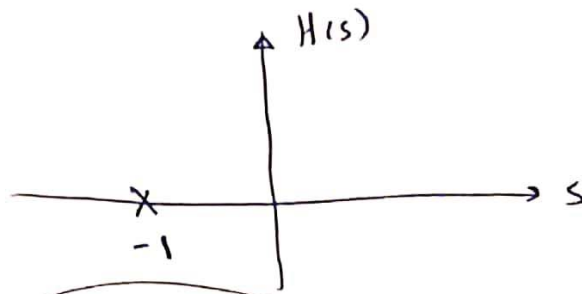
N is the order "specified according to the specifications in frequency domain of LPF"
 ω_c, ω_r, H_r

$$[N=1]$$

$$H(s) \cdot H(-s) = \frac{1}{1-s^2} = \frac{1}{(1-s)(1+s)}$$

This pole is refused as it is not stable

[Poles in LHS \Rightarrow stable]
Poles in RHS \Rightarrow unstable]



$$\therefore H(s) = \frac{1}{s+1} \quad \text{For } N=1$$

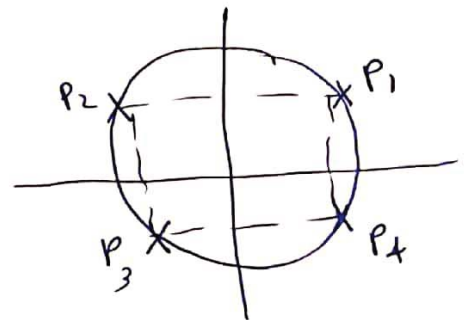
$$[N=2]$$

$$H(s) \cdot H(-s) = \frac{1}{1+s^4}$$

angle $\frac{\pi}{4}$ between each pole

Four poles: P_1, P_2, P_3, P_4

[P_1, P_4 are refused as they are not stable]



$$P_2 = -\frac{1}{\sqrt{2}} + j\frac{1}{\sqrt{2}}, \quad P_3 = -\frac{1}{\sqrt{2}} - j\frac{1}{\sqrt{2}}$$

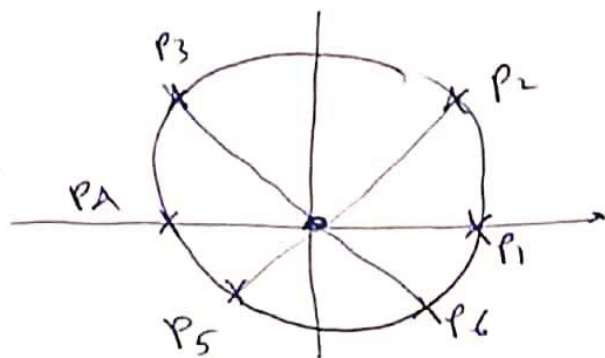
$$H(s) = \frac{1}{(s-P_2)(s-P_3)} = \frac{1}{(s-(-\frac{1}{\sqrt{2}} + j\frac{1}{\sqrt{2}}))(s-(-\frac{1}{\sqrt{2}} - j\frac{1}{\sqrt{2}}))}$$

$$H(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$$

$$N=3 \quad H(s). \quad H(-s) = \frac{1}{1-s^6}$$

Six poles: $p_1, p_2, p_3, p_4, p_5, p_6$

[p_1, p_2, p_6 are not stable]



$\frac{\pi}{3}$ between each pole.

$$H(s) = \frac{1}{(s-p_3)(s-p_4)(s-p_5)}$$

Where $p_3 = -\frac{1}{2} + j\frac{\sqrt{3}}{2}$

$p_4 = -1$

$p_5 = -\frac{1}{2} - j\frac{\sqrt{3}}{2}$

$$\Rightarrow H(s) = \frac{1}{s^3 + 2s^2 + 2s + 1}$$

Summary

Butterworth Approximation of normalized LPF

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

$$N=2 \Rightarrow H(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$$

$$N=3 \Rightarrow H(s) = \frac{1}{s^3 + 2s^2 + 2s + 1}$$

done

Note

N : order of Butterworth Filter

[It should be known from Frequency Specifications of ω_c, ω_r, H_r]

$$|H(\omega)|^2 = \frac{1}{1 + \left(\frac{\omega}{\omega_c}\right)^{2N}}$$

Given: ω_r & H_r beside ω_c

$$\therefore H_r^2 = \frac{1}{1 + \left(\frac{\omega_r}{\omega_c}\right)^{2N}} \Rightarrow \text{get } N = 4$$

Now, we have the normalized Analog LPF transfer function using Butterworth approximation:

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

$$N=2 \Rightarrow H(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$$

Q: We need to transform from analog Domain to Digital Domain [3 methods]

Bilinear
Transformation

Impulse
invariance

Step
invariance

I Bi Linear Transformation

Put $\left[s = \frac{2}{T_s} \frac{z-1}{z+1} \right]$ in $H(s)$ To get $H(z)$

$$\text{i.e. : } H(z) = H(s) \Big|_{s = \frac{2}{T_s} \frac{z-1}{z+1}}$$

also Digital to analog Transformation is a

mapping from analog frequency to digital frequency

$$\left[\omega_a = \frac{2}{T_s} \tan \left(\frac{\omega_d T_s}{2} \right) \right] \text{ "Derivation in next page"}$$

where ω_a : analog frequency

ω_d : digital frequency

T_s : time between 2 successive samples

in time domain $= \frac{1}{f_s}$ ← sampling frequency

$$\rightarrow \text{Recall : } e^{jx} - e^{-jx} = 2j \sin x$$

$$e^{jx} + e^{-jx} = 2 \cos x$$

Q:

Discuss mapping between analog & Digital Frequency in Bilinear Transformation and prove the relation?

Sol - "prove equation relating analog and digital Freq in Bilinear Transformation"

Steps

1 - in Bilinear Transformation
$$S = \frac{2}{T_s} \frac{Z - 1}{Z + 1}$$

2 - but S : For Analog Domain $S = j\omega_a$ ^{analog Freq}

Z : For Digital Domain $Z = e^{j\omega_d T_s}$ ^{digital Freq}

$$j\omega_a = \frac{2}{T_s} \frac{e^{j\omega_d T_s} - 1}{e^{j\omega_d T_s} + 1}$$

$$j\omega_a = \frac{2}{T_s} \cdot \frac{e^{j\omega_d \frac{T_s}{2}}}{e^{j\omega_d \frac{T_s}{2}}} \left[\frac{e^{j\omega_d \frac{T_s}{2}} - e^{-j\omega_d \frac{T_s}{2}}}{e^{j\omega_d \frac{T_s}{2}} + e^{-j\omega_d \frac{T_s}{2}}} \right]$$

$$j\omega_a = \frac{2}{T_s} \frac{2j \sin(\omega_d \frac{T_s}{2})}{2 \cos(\omega_d \frac{T_s}{2})}$$

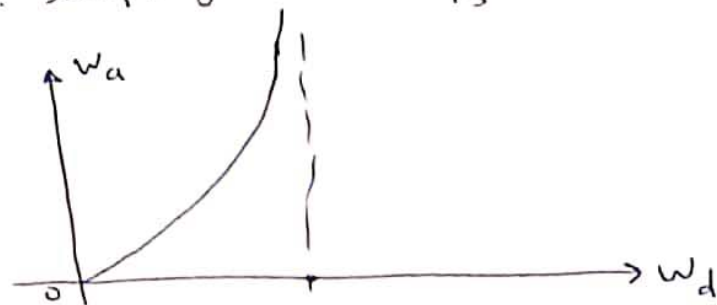
$$j\omega_a = \frac{2j}{T_s} \frac{\sin(\omega_d T_s/2)}{\cos(\omega_d T_s/2)}$$

$$\omega_a = \frac{2}{T_s} \tan\left(\frac{\omega_d T_s}{2}\right) \quad \leftarrow \text{Non linear relation}$$

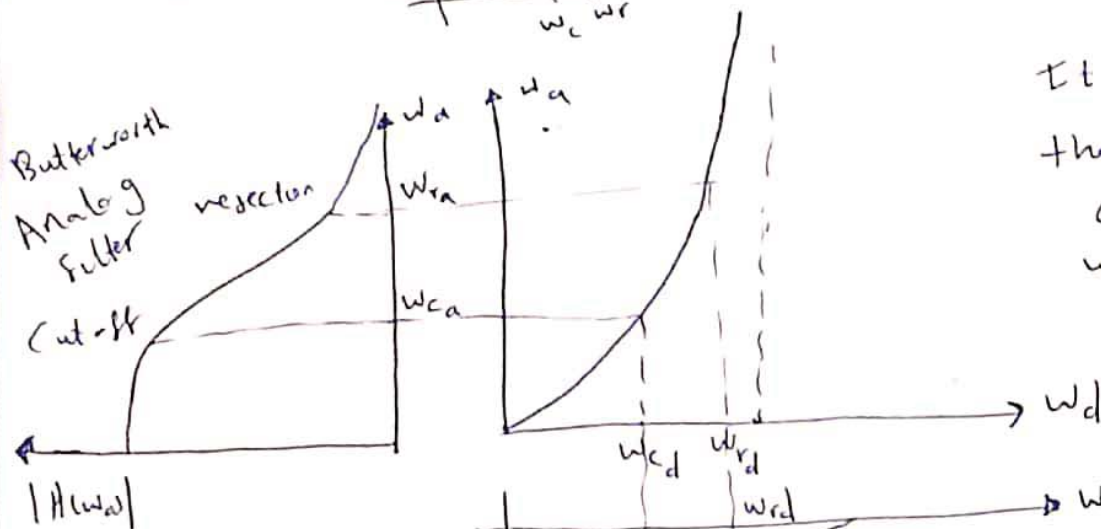
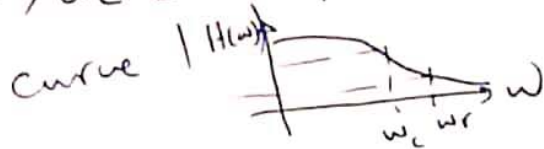
Where: ω_a : analog frequency

ω_d : digital frequency

T_s : Sampling time = $\frac{1}{f_s}$



Now, we can map the analog butterworth approximation



It can be seen that the approximation curve is preserved when transferring from analog to Digital..



Procedures For design of IIR digital Filter using BiLinear transformation :-

Steps

- 1- Find $H(s)$ For anormalized LPF كفوف
 Butterworth $N=1 \Rightarrow H(s) = \frac{1}{s+1}$
 $N=2 \Rightarrow H(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$
- 2- Cut off Frequency of digital Filter $\omega_c = ? \rightarrow$ given $[\omega_c = 2\pi f_c]$

- 3- Find equivalent analog cut off Frequency

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

- 4- Denormalize analog Transfer Function $H(s)$ as Follow: [Convert normalized LPF to denormalized filter]

$$\textcircled{1} \quad \text{LPF}_{\text{normalized}} \longrightarrow \text{LPF}_{\text{denormalized}} : s \longrightarrow \frac{s}{\omega_{ca}}$$

$$\textcircled{2} \quad \text{LPF}_{\text{normalized}} \longrightarrow \text{HPF}_{\text{denormalized}} : s \longrightarrow \frac{\omega_{ca}}{s}$$

$$(3) \text{ LPF}_{\text{normalized}} \longrightarrow \text{BPF}_{\text{denormalized}} \xrightarrow{S} \frac{S^2 + \omega_0^2}{S(B.\omega)}$$

$$(4) \text{ LPF}_{\text{normalized}} \longrightarrow \text{BSF}_{\text{denormalized}} \xrightarrow{S} \frac{S(B.\omega)}{S^2 + \omega_0^2}$$

$$\omega_0 = \sqrt{\omega_c \omega_{cL}}, B\omega = \omega_c - \omega_{cL}$$

5 - Apply BiLinear Transformation

$$S = \frac{2}{T_s} \cdot \frac{z-1}{z+1}$$

$$H(z) = H(s) \Big|_{S = \frac{2}{T_s} \cdot \frac{z-1}{z+1}}$$

then we get Transfer Function For

Digital Filter $H(z)$

$$H(z) = \frac{a_0 + a_1 z^{-1} + a_2 z^{-2} - \dots}{1 + b_1 z^{-1} + b_2 z^{-2} - \dots}$$

(6) Implementation

- Direct Form I
- I / II
- Cascade
- Parallel

Example (1)

Design IIR Digital Filter of order = 1

using Bilinear Transformation. The Filter

is LPF with $f_c = 30 \text{ Hz}$ & $f_s = 150 \text{ Hz}$
_{cut off}

Solution

Note that: the Given cut off Frequency is the digital frequency as the specifications usually are given for the required Digital Filter.

$$\therefore f_{cd} = 30 \text{ Hz}, f_s = 150 \text{ Hz} \Rightarrow T_s = \frac{1}{150}$$

Steps:

$$\boxed{1} \quad \underline{N=1} \Rightarrow H(s) \Big|_{\substack{\text{LPF} \\ \text{normalized}}} = \frac{1}{s+1}$$

$$\boxed{2} \quad \omega_{cd} = 2\pi f_{cd} = 2\pi * 30 = \boxed{60\pi}$$

$$\boxed{3} \quad \omega_c = \frac{2}{T_s} \tan\left(\frac{\omega_{cd} T_s}{2}\right) = \frac{2}{T_s} \tan\left(\frac{60\pi(\frac{1}{150})}{2}\right)$$

$$\omega_c = \frac{2}{T_s} \tan\left(\frac{\pi}{5}\right) = \frac{2}{T_s} \tan(36^\circ) = \boxed{\frac{2}{T_s} 0.72}$$

④

$$H(s) \Big|_{\text{LPF normalized}} = \frac{1}{s+1}$$

$$\downarrow \quad s \rightarrow \frac{s}{\omega_{ca}}$$

$$H(s) \Big|_{\text{LPF denormalized}} = \quad s \rightarrow \frac{s}{\omega_{ca}}$$

$$H(s) \Big|_{\text{Denormalized}} = \frac{1}{\left(\frac{s}{\omega_{ca}}\right) + 1} = \frac{\omega_{ca}}{s + \omega_{ca}}$$

$$H(s) \Big|_{\text{Denormalized}} = \frac{\frac{2}{T_s} (0.72)}{s + \frac{2}{T_s} 0.72}$$

⑤ Bilinear Transformation:

put $s = \frac{2}{T_s} \frac{z-1}{z+1}$ in $H(s) \Big|_{\text{Denormalized}}$

$$H(z) = \frac{\left(\frac{2}{T_s}\right) 0.72}{\left(\frac{2}{T_s}\right) \frac{z-1}{z+1} + \left(\frac{2}{T_s}\right) 0.72} = \frac{0.72}{\frac{z-1}{z+1} + 0.72}$$

$$H(z) = \frac{0.72}{\frac{z-1}{z+1} + 0.72} \times \frac{z+1}{z+1}$$

$$H(z) = \frac{0.72z + 0.72}{z-1 + 0.72z + 0.72}$$

$$H(z) = \frac{0.72 + 0.72z^{-1}}{1.72z - 0.28} = \frac{0.72 + 0.72z^{-1}}{1.72 - 0.28z^{-1}}$$

↑
should be (1)

Before implementation

↓
Coefficient of $y(n)$ must be (1)

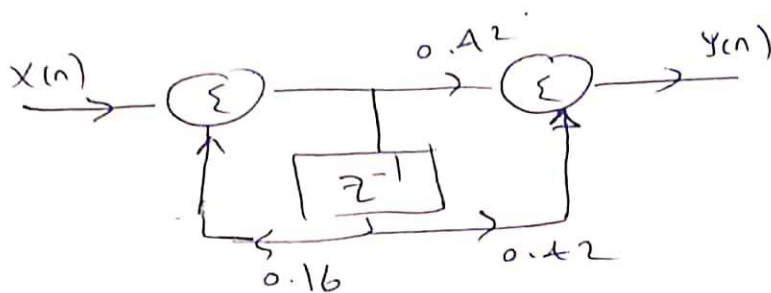
Divide by 1.72

$$H(z) = \frac{0.42 + 0.42z^{-1}}{1 - 0.16z^{-1}}$$

[6] Implementation (Realization)

Direct Form I or Direct Form II

Let's make it using Direct Form II



Example (2)

Design using Bilinear Transformation a 2nd order

HPF digital IIR filter having $f_c = 2 \text{ KHz}$
& $f_s = 6 \text{ KHz}$

Solution

Given: $f_c = 2 \text{ KHz}$, $T_s = \frac{1}{f_s} = \frac{1}{6 \times 10^3}$

Steps:

[1] $N=2 \Rightarrow H(s) \Big|_{\text{LPF normalized}} = \frac{1}{s^2 + \sqrt{2}s + 1}$

[2] $\omega_{cd} = 2\pi f_{cd} = 2\pi \times 2 \times 10^3 = 4\pi \times 10^3$

[3] $\omega_{ca} = \frac{2}{T_s} \tan\left(\frac{\omega_{cd} T_s}{2}\right) = \frac{2}{T_s} \tan\left(\frac{4\pi \times 10^3 \times \frac{1}{6 \times 10^3}}{2}\right)$

$\omega_{ca} = \frac{2}{T_s} \tan\left(\frac{\pi}{6}\right) = \frac{2}{T_s} \sqrt{3}$, $T_s = \frac{1}{6 \times 10^3}$

[4] $H(s) \Big|_{\text{LPF normalized}} = \frac{1}{s^2 + \sqrt{2}s + 1}$
 $\Downarrow \xrightarrow{s \rightarrow \frac{\omega_{ca}}{s}}$

$H(s) \Big|_{\text{HPF Denormalized}} = \frac{s}{s^2 + \omega_{ca}}$

$$H(s) \Big|_{\text{Denormalized}} = \frac{1}{\left(\frac{\omega_{ca}}{s}\right)^2 + \sqrt{2} \left(\frac{\omega_{ca}}{s}\right) + 1} \quad * \frac{s^2}{s^2}$$

$$= \frac{s^2}{s^2 + \sqrt{2} \omega_{ca} s + \omega_{ca}^2} \quad \text{but } \omega_{ca} = \frac{2}{T_s} \sqrt{3}$$

$$H(s) \Big|_{\text{Denormalized}} = \frac{s^2}{s^2 + \frac{2}{T_s} \sqrt{6} s + \left(\frac{2}{T_s}\right)^2 3}$$

5 Bilinear Transformation:

$$\text{put } s = \frac{2}{T_s} \frac{z-1}{z+1}$$

$$\therefore H(z) = \frac{\left(\frac{2}{T_s}\right)^2 \left(\frac{z-1}{z+1}\right)^2}{\left(\frac{2}{T_s}\right)^2 \left(\frac{z-1}{z+1}\right)^2 + \frac{2}{T_s} \sqrt{6} \frac{2}{T_s} \cdot \frac{z-1}{z+1} + \left(\frac{2}{T_s}\right)^2 3}$$

$$H(z) = \frac{\cancel{\left(\frac{2}{T_s}\right)^2} \frac{(z-1)^2}{(z+1)^2}}{\cancel{\left(\frac{2}{T_s}\right)^2} \left(\frac{z-1}{z+1}\right)^2 + \cancel{\left(\frac{2}{T_s}\right)^2} \sqrt{6} \frac{z-1}{z+1} + 3 \cancel{\left(\frac{2}{T_s}\right)^2}}$$

$$H(z) = \frac{\frac{(z-1)^2}{(z+1)^2}}{\frac{(z-1)^2}{(z+1)^2} + \sqrt{6} \frac{z-1}{z+1} + 3} \quad * \frac{(z+1)^2}{(z+1)^2}$$

$$H(z) = \frac{(z-1)^2}{(z-1)^2 + \sqrt{6}(z^2-1) + 3(z+1)^2}$$

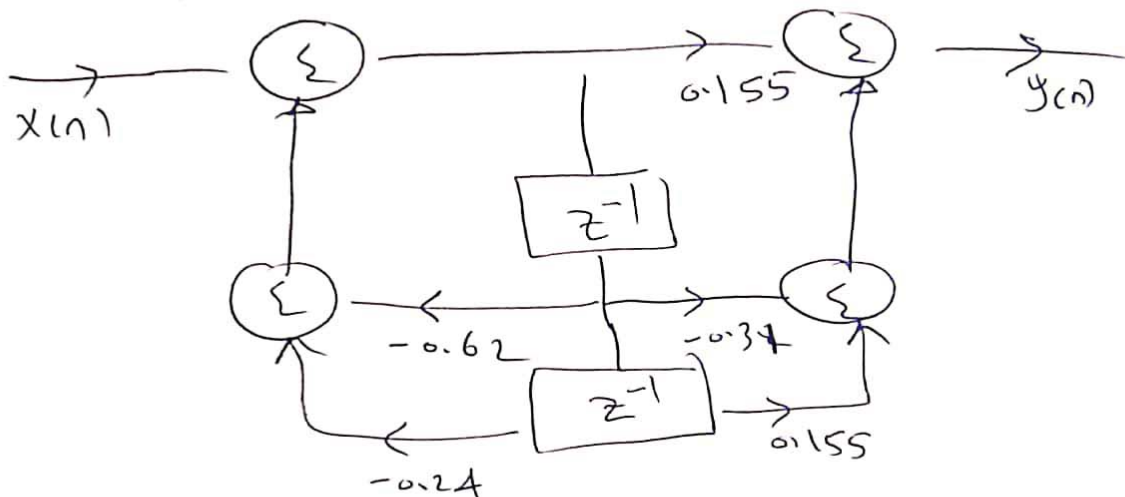
$$H(z) = \frac{z^2 - 2z + 1}{6.45z^2 + 4z + 1.55} = \frac{1 - 2z^{-1} + z^{-2}}{6.45 + 1z^{-1} + 1.55z^{-2}}$$

Divide by (6.45) as Gain of $y(n)$ should be 1

$$H(z) = \frac{0.155 - 0.31z^{-1} + 0.155z^{-2}}{1 + 0.62z^{-1} + 0.24z^{-2}}$$

[6] Make the realization $\left\{ \begin{array}{l} \text{Direct Form I} \\ \text{or} \\ \text{Direct Form II} \end{array} \right.$

let's make it Direct Form II



Example (3)

Using Bilinear transformation, Design IIR Digital
HPF Filter having cut off 1 KHz, Sampling Frequency = 4 KHz
The Filter order = 3

Solution

Given: $f_c = 1 \text{ KHz}$, $f_s = 4 \text{ KHz}$, $T_s = \frac{1}{4 \times 10^3}$

Steps:

[1] $N = 3 \Rightarrow H(s) = \frac{1}{s^3 + 2s^2 + 2s + 1}$
LPF normalized

[2] $\omega_{cd} = 2\pi f_{cd} = 2\pi(1\text{K}) = 2\pi \times 10^3$

[3] $\omega_{ca} = \frac{2}{T_s} \tan\left(\frac{\omega_{cd} T_s}{2}\right)$

$$\omega_{ca} = \frac{2}{T_s} \tan\left(\frac{2\pi \times 10^3 \times \frac{1}{4 \times 10^3}}{2}\right)$$

$$\omega_{ca} = \frac{2}{T_s} \tan\left(\frac{\pi}{4}\right) = \left(\frac{2}{T_s} \cdot \frac{1}{\sqrt{2}}\right)$$

[4] LPF normalized $\xrightarrow[\omega_{ca}]{s}$ HPF denormalized

$$H(s) \Big|_{\text{Denormalized}} = \frac{1}{s^3 + 2s^2 + 2s + 1} \quad s = \frac{\omega_{ca}}{s}$$

$$H(s) \Big|_{\text{Denormalized}} = \frac{1}{\frac{\omega_{ca}^3}{s^3} + 2 \frac{\omega_{ca}^2}{s^2} + 2 \frac{\omega_{ca}}{s} + 1} \quad * \frac{s^3}{s^3}$$

$$H(s) \Big|_{\text{Denormalized}} = \frac{s^3}{\omega_{ca}^3 + 2 \omega_{ca}^2 s + 2 \omega_{ca} s^2 + s^3}$$

$$H(s) \Big|_{\text{Denormalized}} = \frac{s^3}{\left(\frac{2}{T_s}\right)^3 \left(\frac{1}{r_2}\right)^3 + 2 \left(\frac{2}{T_s}\right)^2 \frac{1}{2} s + 2 \left(\frac{2}{T_s}\right) \frac{1}{r_2} s^2 + s^3}$$

5 Bilinear Transformation

put $s = \frac{2}{T_s} \frac{z-1}{z+1}$ in $H(s) \Big|_{\text{Denormalized}}$

$$H(z) = \frac{\left(\frac{2}{T_s}\right)^3 \left(\frac{z-1}{z+1}\right)^3}{\left(\frac{2}{T_s}\right)^3 \frac{1}{(2)^{3/2}} + \frac{2}{2} \left(\frac{2}{T_s}\right)^3 \left(\frac{z-1}{z+1}\right) + \frac{2}{\sqrt{2}} \left(\frac{2}{T_s}\right)^3 \left(\frac{z-1}{z+1}\right)^2 + \left(\frac{2}{T_s}\right)^3 \left(\frac{z-1}{z+1}\right)^3}$$

$$H(z) = \frac{(z-1)^3 / (z+1)^3}{\frac{1}{(2)^{3/2}} + \frac{(z-1)}{(z+1)} + \frac{2}{\sqrt{2}} \frac{(z-1)^2}{(z+1)^2} + \frac{(z-1)^3}{(z+1)^3}}$$

multiply by $\frac{(z+1)^3}{(z+1)^3}$

$$H(z) = \frac{(z-1)^3}{\frac{1}{(2)^{3/2}} (z+1)^3 + (z-1)(z+1)^2 + \frac{2}{\sqrt{2}} (z-1)^2(z+1) + (z-1)^3}$$

After simplifying

$$H(z) = \frac{z^3 - 3z^2 + 3z - 1}{3.4z^3 - 3.4z^2 + 3.4z - 3.4}$$

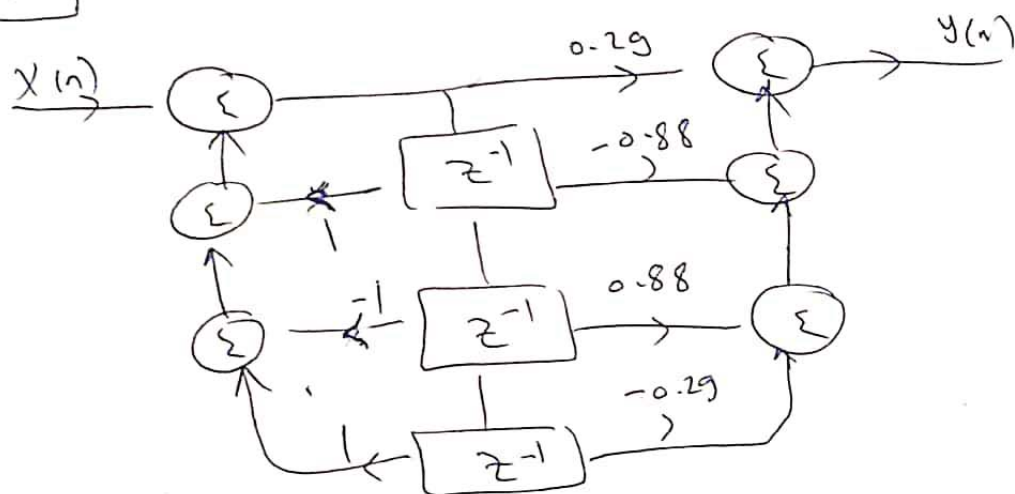
$$H(z) = \frac{1 - 3z^{-1} + 3z^{-2} - z^{-3}}{3.4 - 3.4z^{-1} + 3.4z^{-2} - 3.4z^{-3}}$$

This should be (1)

Divide by 3.4

$$H(z) = \frac{0.29 - 0.98z^{-1} + 0.88z^{-2} - 0.29z^{-3}}{1 - z^{-1} + z^{-2} - z^{-3}}$$

6 Implement \Rightarrow Direct Form II



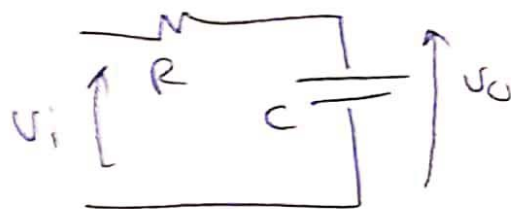
Note

بہ ایں

What is the meaning of Denormalization?
in Step (4)

Ex:

RC LPF

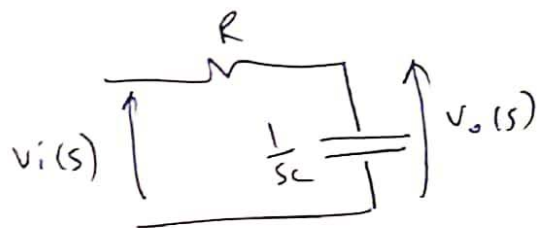


LPF normalized \rightarrow LPF Denorm
 \rightarrow HPF Denorm
 \rightarrow BPF Denorm
 \rightarrow BSF Denorm

Find $H(s)$ of LPF $\left\{ \begin{array}{l} \text{normalized [put } R=C=1] \\ \text{Denormalized in terms of} \\ \text{values of } R, C \end{array} \right.$

SOL

$$V_o(s) = V_i(s) \cdot \frac{\frac{1}{sC}}{\frac{1}{sC} + R}$$



$$\frac{V_o(s)}{V_i(s)} = H(s) = \frac{\frac{1}{sC}}{\frac{1}{sC} + R} = \frac{1}{1 + RCs}$$

$$H(s) \Big|_{\text{Denormalized}} = \frac{1}{1 + RCs}$$

$$\omega_c = \frac{1}{RC}$$

$$\rightarrow H(s) \Big|_{\text{normalized } R=1, C=1} = \frac{1}{s+1} \quad [\text{As we know}]$$

to convert from LPF normalized $\xrightarrow{s \rightarrow \frac{s}{\omega_c}}$ LPF denormalized

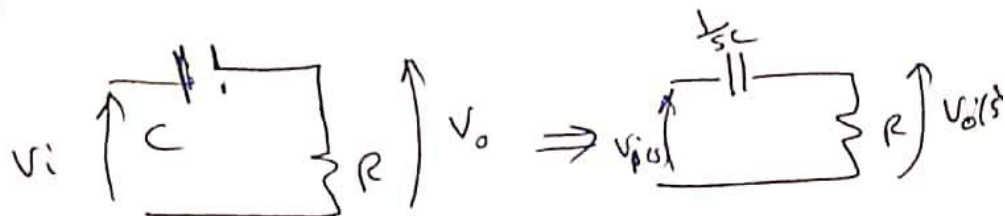
let's check

$$s \rightarrow \frac{s}{\omega_c} = \frac{s}{\frac{1}{RC}} = RCS$$

$$\left. H(s) \right|_{\text{Denormalized}} = \frac{1}{s+1} \xrightarrow{s \rightarrow RCS} RCS$$

$$\left(\left. H(s) \right|_{\text{Denormalized}} = \frac{1}{1+RCS} \right) \leftarrow \text{the same as we got}$$

→ For HPF



$$V_o(s) = V_i(s) \cdot \frac{R}{R + \frac{1}{sC}} \Rightarrow \left(\left. H(s) \right|_{\text{Denormalized}} = \frac{sCR}{1+RCS} \right)$$

$$\omega_c = \frac{1}{RC}$$

But For $[N=1] \Rightarrow$ LPF normalized $H(s) = \frac{1}{1+s}$

Convert From LPF normalized $\xrightarrow{s \xrightarrow{\omega_c} \frac{s}{s}} \frac{s}{s} \rightarrow$ HPF Denormalized

$$\left. H(s) \right|_{\text{HPF Denormalized}} = \frac{1}{1+s} \xrightarrow{s \rightarrow \frac{1}{RCS}} = \frac{1}{1 + \frac{1}{RCS}} = \frac{RCS}{1+RCS}$$

and so on ✓

←← The same as we got