

Digital Signal Processing

IIR Filter Design

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NCL

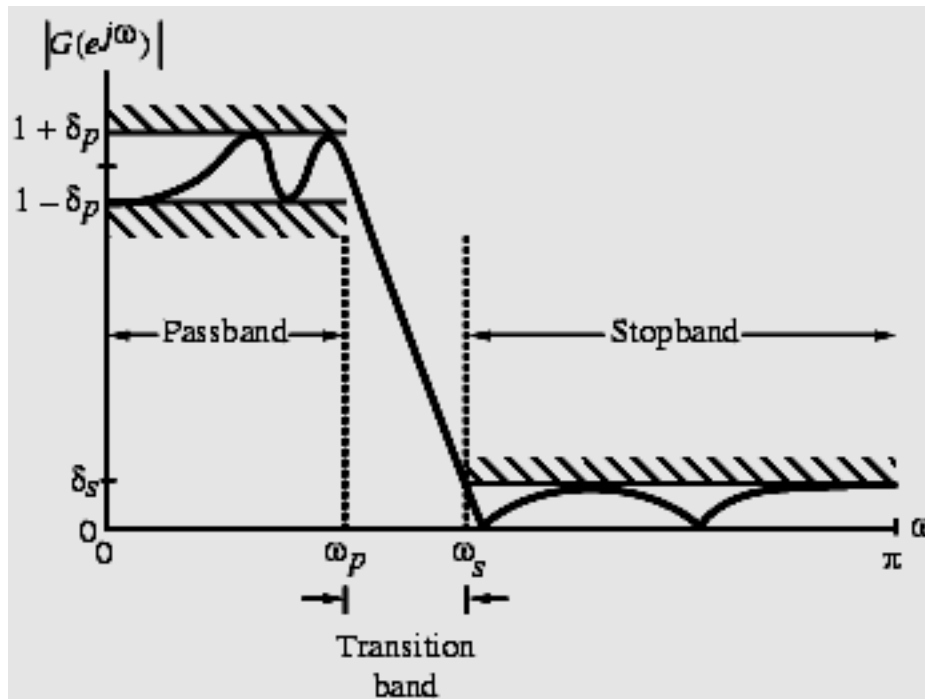
UESTC

Filter Specification

Example: LPF

$$|G(e^{j\omega})| \leq \delta_s \quad \text{for } \omega_s \leq |\omega| \leq \pi$$

$$1 - \delta_P \leq |G(e^{j\omega})| \leq 1 + \delta_P \quad \text{for } |\omega| \leq \omega_P$$



Passband : ω_P

Stopband : ω_S

Filter Selection

The *objective* of filter design is to design casual $H(z)$ that meets the specifications

IIR or FIR ?

IIR:

low order to reduce
computational complexity,
but may not stable.

FIR:

with linear phase ,but high order

$$H[n] = \pm h[N - n]$$

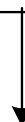
$$H(z) = \frac{\sum_{k=0}^M p_k z^{-k}}{\sum_{k=0}^N d_k z^{-k}}$$
$$H[z] = \sum_{k=0}^N h[k] z^{-k}$$

IIR Design

IIR design

The most common practice is to transform $H_a(s)$ into the desired digital transfer function $G(z)$.

convert the digital filter specifications to
into analog lowpass prototype filter
specifications



determine the analog lowpass filter
transfer function meeting these
specifications



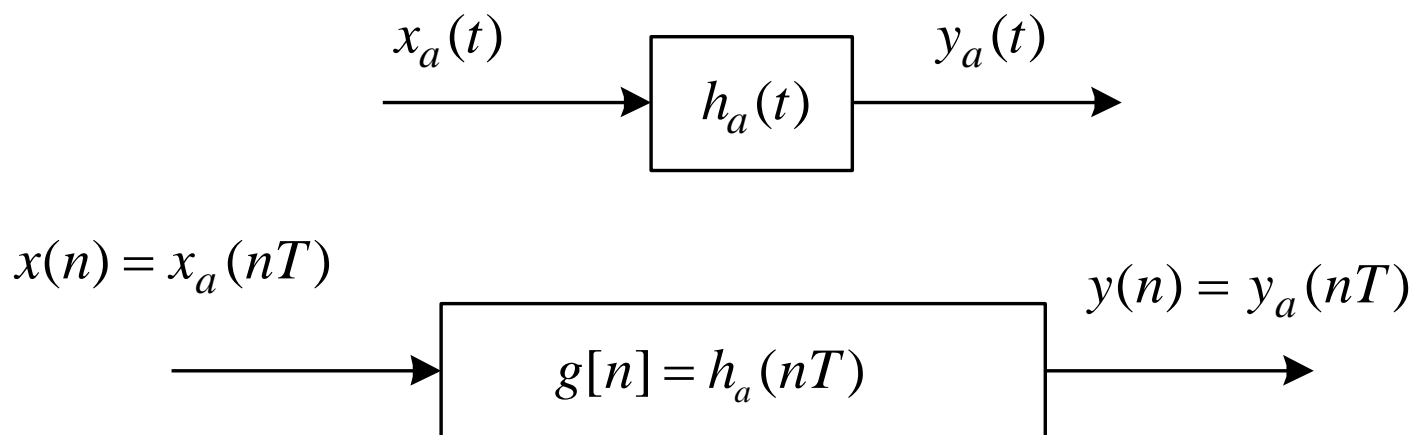
transform the analog filter into the
desired digital transfer function

IIR Design

- Apply a mapping from s -domain to the z -domain so that the essential properties of the analog frequency response are preserved.
- The mapping function should:
 - (1) The **imaginary axis** in the s -plane be mapped onto the **unit circle** of the z -plane.
 - (2) A **stable** analog transfer function be transferred into a **stable** digital transfer function.

IIR Design

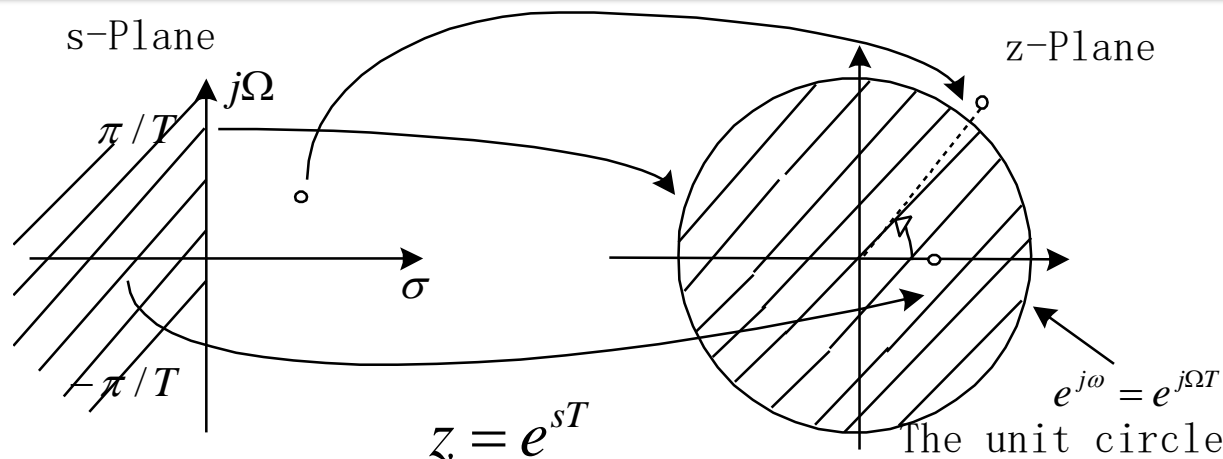
Impulse Invariance Method



$$G(z)|_{z=e^{sT}} = \frac{1}{T} \sum_{m=-\infty}^{\infty} H_a\left(s - jm \frac{2\pi}{T}\right)$$

The mapping function $z = e^{sT}$

Impulse Invariance Method



- $j\Omega$ -axis mapped onto the unit circle, stable region in s-plane mapped into the stable region in z-plane).
- $j\Omega$ -axis in the s-plane and the unit circle in $\omega = \Omega T$
- Nyquist

$$G(e^{j\omega}) = \frac{1}{T} H_a(j\Omega) \quad \text{for } |\omega| < \pi$$

Design Steps

$H_a(s)$ is partial-fractional expressed as: ($h_a(t)$ is derived)

$$H_a(s) = \sum_{i=1}^N \frac{A_i}{s - s_i} \quad \text{Re}(s_i) < 0$$

(2) From $g[n] = h_a(nT)$

$$G_1(z) = \sum_{i=1}^N \frac{A_i}{1 - e^{s_i T} z^{-1}}$$

$$G(z) = \sum_{i=1}^N \frac{TA_i}{1 - e^{s_i T} z^{-1}}$$

Proof

$$h_a(t) = \mathcal{L}^{-1}\{H_a(s)\} = \sum_{i=1}^N A_i e^{s_i t} u(t)$$

$$g[n] = h_a(nT) = \sum_{i=1}^N A_i e^{s_i nT} u[n]$$

$$G_1(z) = \sum_{i=1}^N \frac{A_i}{1 - e^{s_i T} z^{-1}}$$

Example

$$H_a(s) = \frac{2}{s^2 + 3s + 2}$$

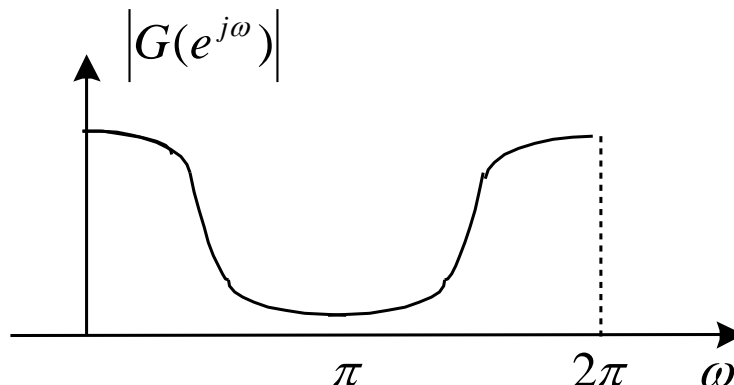
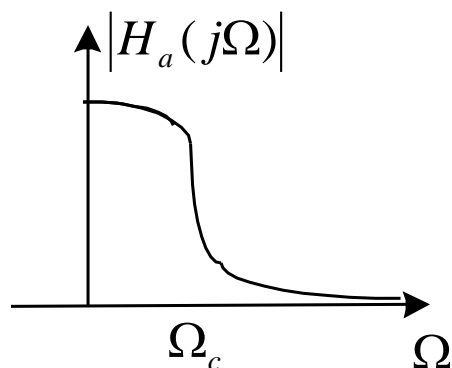


$$H_a(s) = \frac{2}{s+1} - \frac{2}{s+2}$$

$$G(z) = T \frac{2}{1 - e^{-T} z^{-1}} - T \frac{2}{1 - e^{-2T} z^{-1}} = \frac{2T(e^{-T} - e^{-2T})z^{-1}}{1 - (e^{-2T} + e^{-T})z^{-1} + e^{-3T}z^{-2}}$$

$$G(z) = \frac{0.4651 z^{-1}}{1 - 0.5032 z^{-1} + 0.04979 z^{-2}}$$

When $T = 1$



Bilinear Translation Method

map from s-plane
to z-plane

$$s = \frac{2}{T} \left(\frac{1 - z^{-1}}{1 + z^{-1}} \right)$$

$$G(z) = H_a(s) \Big|_{s = \frac{2}{T} \left(\frac{1 - z^{-1}}{1 + z^{-1}} \right)}$$

For $s = \sigma_0 + j\Omega_0$

$$z = \frac{1 + s}{1 - s} = \frac{1 + \sigma_0 + j\Omega_0}{1 - \sigma_0 + j\Omega_0}$$

$$\sigma_0 = 0 \quad |z| = 1$$

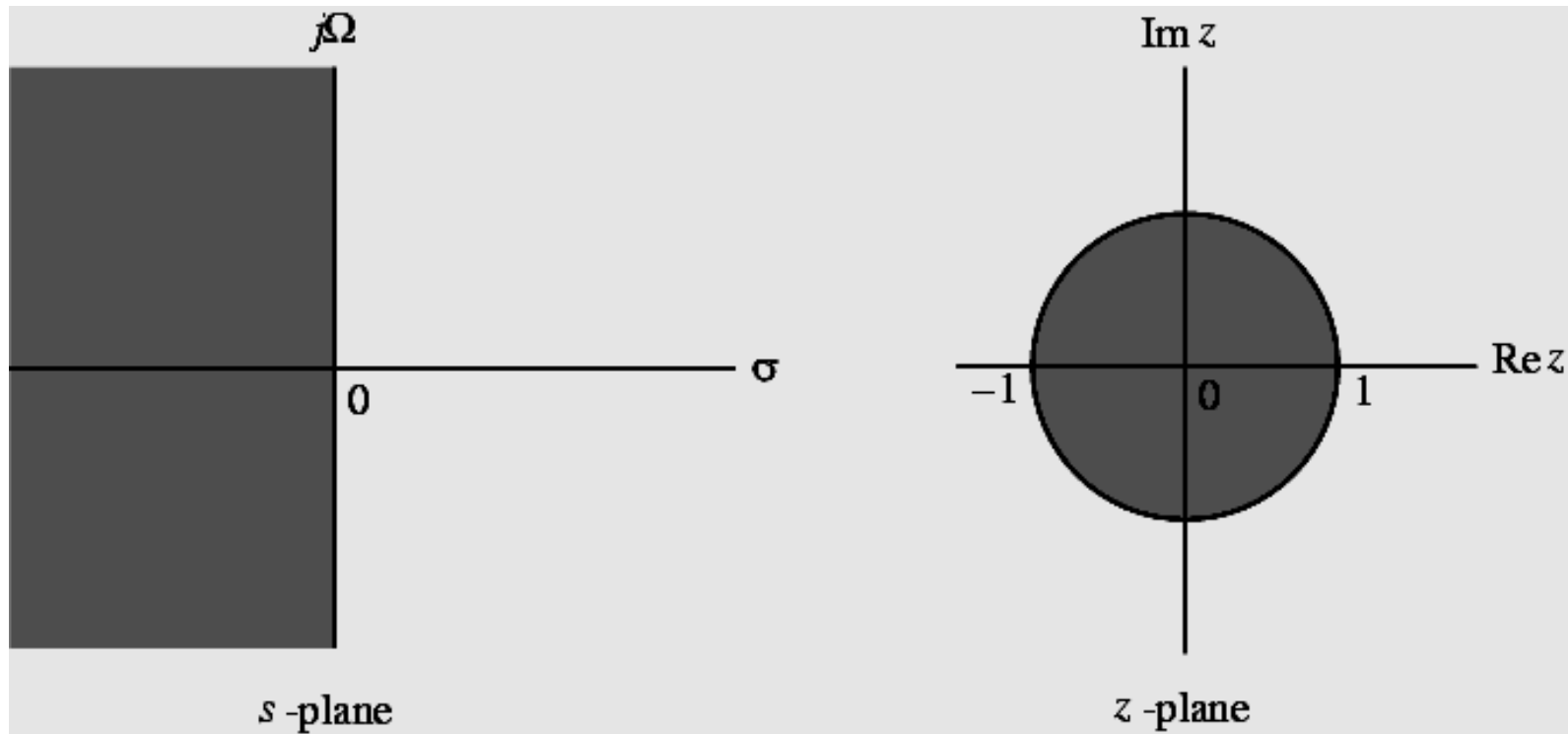
$$\sigma_0 < 0 \quad |z| < 1$$

$$\sigma_0 > 0 \quad |z| > 1$$

$$|z|^2 = \frac{(1 + \sigma_0)^2 + (\Omega_0)^2}{(1 - \sigma_0)^2 + (\Omega_0)^2}$$

IIR Design

Bilinear Translation Method



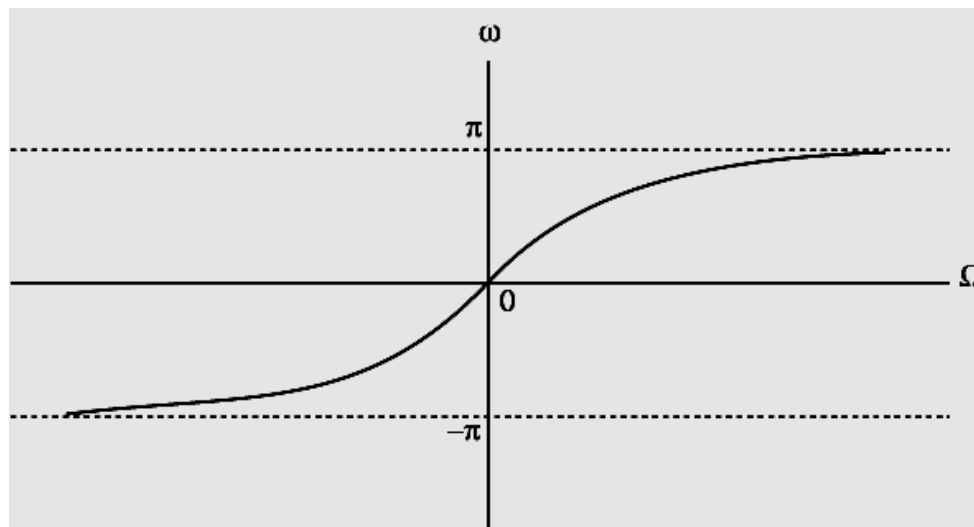
IIR Design

Bilinear Translation Method

- For $z = e^{j\omega}$ with $T = 2$ we have

$$j\Omega = \frac{1 - e^{-j\omega}}{1 + e^{-j\omega}} = \frac{e^{-j\omega/2} (e^{j\omega/2} - e^{-j\omega/2})}{e^{-j\omega/2} (e^{j\omega/2} + e^{-j\omega/2})} = j \tan(\omega/2)$$

$$\Omega = \tan(\omega/2)$$

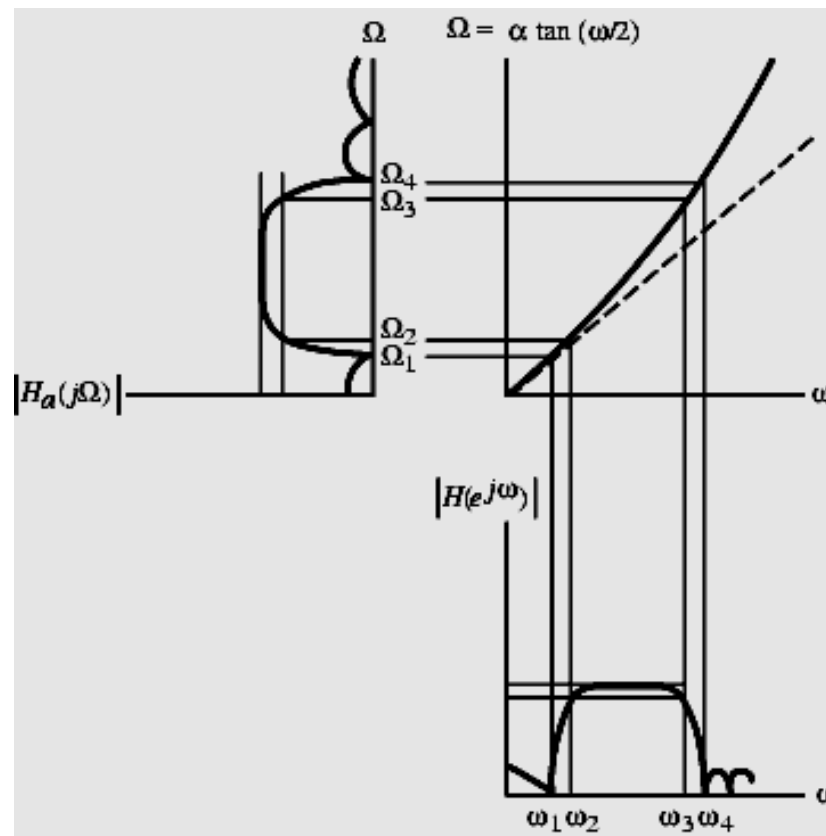


frequency warping

IIR Design

Bilinear Translation Method

Thus, to develop a digital filter meeting a specified magnitude, we must first prewarp the critical band edge frequencies (ω_p and ω_s) to find their analog Equivalents (Ω_s and Ω_p) using Eq.(9.18).



IIR Design

Bilinear Translation Method

- Pre-warp (ω_p, ω_s) to find their analog equivalents (Ω_p, Ω_s)
- Design the analog filter $H_a(s)$
- Design the digital filter $G(z)$ by applying bilinear transformation to $H_a(s)$
- Transformation does not preserve phase response of analog filter

Example

$$H_a(s) = \frac{\Omega_c}{s + \Omega_c}$$

$$G(z) = H_a(s) \Big|_{s = \frac{2}{T} \left(\frac{1-z^{-1}}{1+z^{-1}} \right)} = \frac{\Omega_c (1 - z^{-1})}{(1 - z^{-1}) + \Omega_c (1 + z^{-1})}$$

$$= \frac{1 - \alpha}{2} \frac{1 + z^{-1}}{1 - \alpha z^{-1}}$$

$$\alpha = \frac{1 - \Omega_c}{1 + \Omega_c} = \frac{1 - \tan(\omega_c/2)}{1 + \tan(\omega_c/2)}$$

Example

$$H_a(s) = \frac{s^2 + \Omega_0^2}{s^2 + Bs + \Omega_0^2}$$

$$G(z) = H_a(s) \Big|_{s = \frac{2}{T} \left(\frac{1-z^{-1}}{1+z^{-1}} \right)}$$

$$|H_a(j\Omega_0)| = 0 \quad |H_a(j0)| = |H_a(j\infty)| = 1$$

- Ω_0 is called the **notch frequency**
- If $|H_a(j\Omega_2)| = |H_a(j\Omega_1)| = 1/\sqrt{2}$
 $B = \Omega_2 - \Omega_1$ is the 3-dB notch bandwidth

$$= \frac{(1 + \Omega_0^2) - 2(1 - \Omega_0^2)z^{-1} + (1 + \Omega_0^2)z^{-2}}{(1 + \Omega_0^2 + B) - 2(1 - \Omega_0^2)z^{-1} + (1 + \Omega_0^2 - B)z^{-2}}$$

$$= \frac{1 + \alpha}{2} \frac{1 - 2\beta z^{-1} + z^{-2}}{1 - \beta(1 + \alpha)z^{-1} + \alpha z^{-2}}$$

$$\alpha = \frac{1 + \Omega_0^2 - B}{1 + \Omega_0^2 + B} = \frac{1 - \tan(B_w/2)}{1 + \tan(B_w/2)}$$

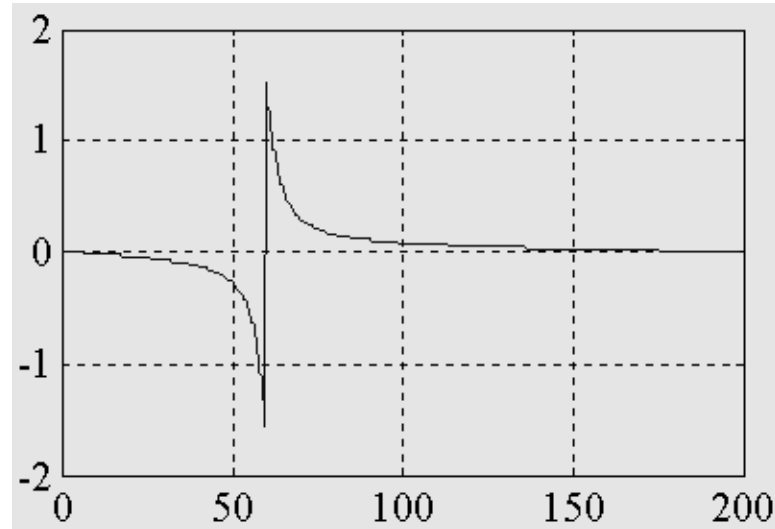
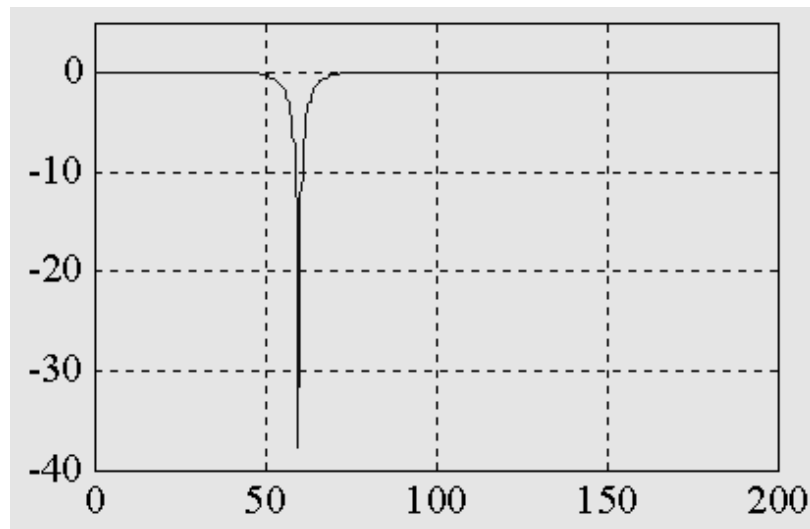
$$\beta = \frac{1 - \Omega_0^2}{1 + \Omega_0^2} = \cos(\omega_0)$$

IIR Design

Example

$$H_a(s) = \frac{s^2 + \Omega_o^2}{s^2 + B s + \Omega_o^2}$$

$$G(z) = \frac{0.954965 - 1.1226287 z^{-1} + 0.954965 z^{-2}}{1 - 1.1226287 z^{-1} + 0.909993 z^{-2}}$$



Case Study

Design $G(z)$ with a maximally flat magnitude, and a pass ripple not exceeding 0.5dB, and the minimum stopband attenuation 15dB.

$$20\log|G(e^{j\omega_p})| > -0.5dB \quad 20\log|G(e^{j\omega_s})| < -15dB$$

$$\omega_p = 0.25\pi, \omega_s = 0.55\pi$$

(1) Prewarping

$$\Omega_P = \tan\left(\frac{\omega_P}{2}\right) = \tan\left(\frac{0.25\pi}{2}\right) = 0.4142136$$

$$\Omega_S = \tan\left(\frac{\omega_S}{2}\right) = \tan\left(\frac{0.55\pi}{2}\right) = 1.1708496$$

(2) Design the parent analog filter $H_a(s)$

From the specifications we obtain

pass ripple 0.5dB



$$20\log(1/\sqrt{1+\varepsilon^2}) = -0.5$$

$$\varepsilon = 0.122$$

stopband attenuation 15dB



$$20\log(1/A) = -15$$

$$A^2 = 31.6228$$

IIR Design

$$k = \frac{\Omega_s}{\Omega_p} = \frac{1.1708496}{0.4142136} = 2.8266809$$

$$k_1 = \frac{\varepsilon}{\sqrt{A^2 - 1}} = 0.0631234$$

$$N = \frac{\log_{10}(1/k_1)}{\log_{10}(1/k)} = \frac{\log_{10}(15.841979)}{\log_{10}(2.8266814)} = 2.6586997$$

The least order of Butterworth LPF is $N = 3$.

IIR Design

the 3-dB frequency

$$\Omega_c = (\varepsilon)^{-1/N} \Omega_p = 1.419915 \Omega_p = 0.588148$$

third-order normalized lowpass Butterworth
transfer function

$$H_{an}(s) = \frac{1}{(s+1)(s^2+s+1)}$$

which has 3-dB frequency at $\Omega_c = 1$.

$$H_a(s) = H_{an}\left(\frac{s}{\Omega_c}\right) = \frac{0.203451}{(s+0.588148)(s^2+0.588148s+0.245918)}$$

**Digital filter $G(z)$ by applying
bilinear transformation**

$$G(z) = H_a(s) \Big|_{s=\frac{2}{T} \left(\frac{1-z^{-1}}{1+z^{-1}} \right)}$$

IIR Design—The Impulse Invariance Method

The third-order lowpass Butterworth with 3-dB frequency at Ω_c .

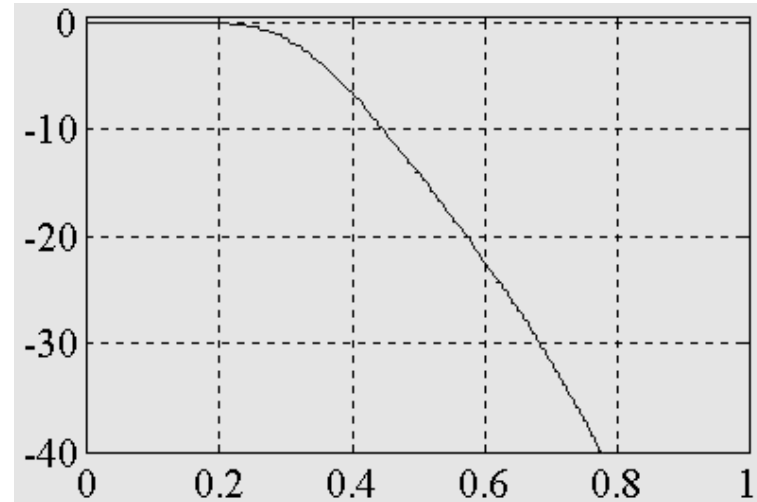
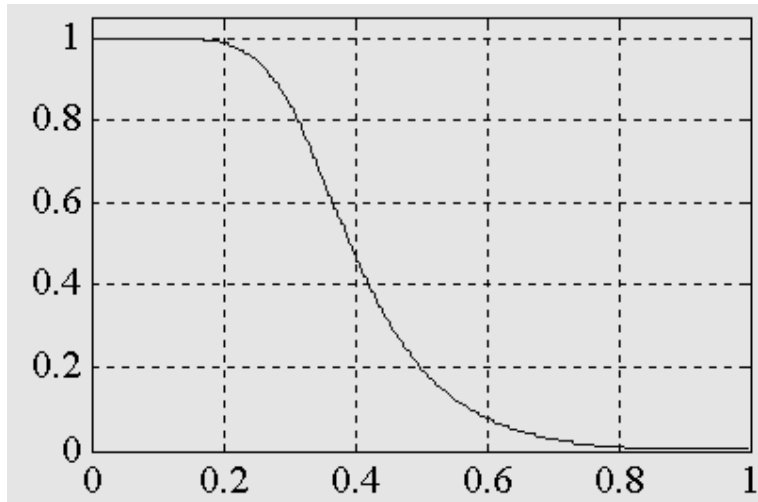
$$H_a(s) = H_{an}\left(\frac{s}{\Omega_c}\right) = \frac{1}{1 + 2\frac{s}{\Omega_c} + 2\left(\frac{s}{\Omega_c}\right)^2 + \left(\frac{s}{\Omega_c}\right)^3}$$

$$\begin{aligned} H_a(s) &= \frac{\Omega_c^3}{(s + \Omega_c)(s - \Omega_c e^{j2\pi/3})(s - \Omega_c e^{-j2\pi/3})} \\ &= \frac{\Omega_c}{s + \Omega_c} + \frac{-(\Omega_c/\sqrt{3})e^{j\pi/6}}{s + \Omega_c(1 - j\sqrt{3})/2} + \frac{-(\Omega_c/\sqrt{3})e^{-j\pi/6}}{s + \Omega_c(1 + j\sqrt{3})/2} \end{aligned}$$

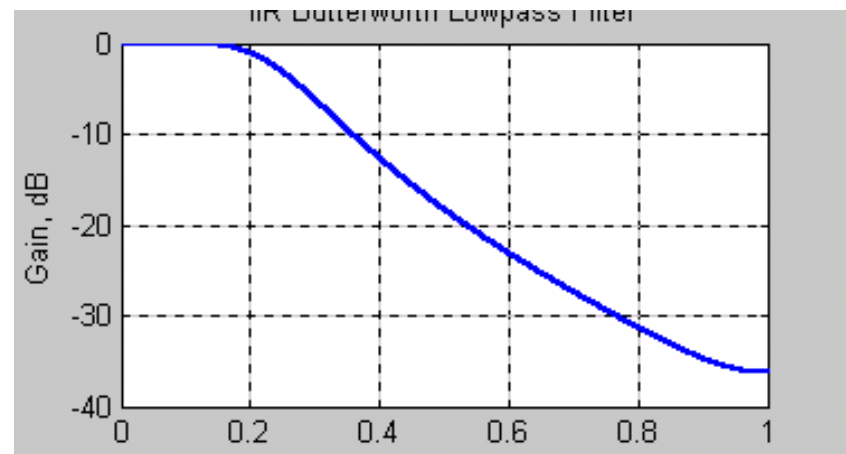
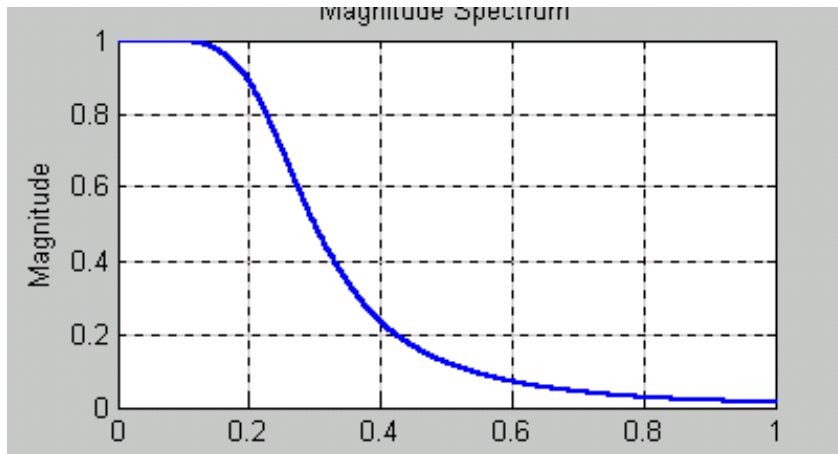
digital transfer function

$$G(z) = \frac{\omega_c}{1 - e^{\omega_c} z^{-1}} + \frac{-(\omega_c/\sqrt{3})e^{j\pi/6}}{1 + e^{-\omega_c(1 - j\sqrt{3})/2} z^{-1}} + \frac{-(\omega_c/\sqrt{3})e^{-j\pi/6}}{1 + e^{\omega_c(1 + j\sqrt{3})/2} z^{-1}} \quad \omega_c = \pi/4$$

IIR Design



The Bilinear Translation Method



The Impulse Invariance Method

IIR Design with Matlab

Read and exercise by yourself!

Note: LPF design **directly in digital form**

1. Order Estimation

```
[N,Wn] = buttord(Wp, Ws, Rp, Rs);
```

2. Filter Design

```
[b,a] = butter(N,Wn);
```

```
% Getting:  $G(z) = B(z) / A(z)$ 
```

3. Other type filter

```
[b,a] = cheby1(N,Rp,Wp,'high');
```

```
%Wn=[W1, W2]
```

IIR Design with Matlab

Note: Digital LPF design in **analog** form by **Bilinear Transform**

1. Order Estimation

```
[N,Wn] = ellipord(Wp,Ws,Rp,Rs,'s');
```

2. Filter Design

```
[bt,at] = ellip(N,Wn);
```

```
%Getting:  $H_a(z) = B_t(z) / A_t(z)$ 
```

3. Bilinear Transform

```
[num,den]= bilinear(b,a,0.5);
```

```
% 0.5 means  $T=2, F_s=0.5$ 
```

```
% Getting:  $G(z) = B(z) / A(z)$ 
```