

MM3: Frequency Transformation of LP-IIR Filters and Linear phase systems

1. Explanation of Exercise Two
2. Frequency transformations
3. Linear phase systems

Explanation of Exercise Two



DT Lowpass IIR Filter Design

Discrete filters

- Butterworth...
- Chebyshev...
- Elliptic ...

Analog filters

- Butterworth...
- Chebyshev...
- Elliptic ...

CT Butterworth lowpass Filters

■ Characteristics

- The magnitude response is maximally flat in the passband
- The magnitude response is monotonic in the passband and stopband

■ Magnitude-squared function

$$|H_c(j\Omega)|^2 = \frac{1}{1 + (j\Omega / j\Omega_c)^{2N}}$$

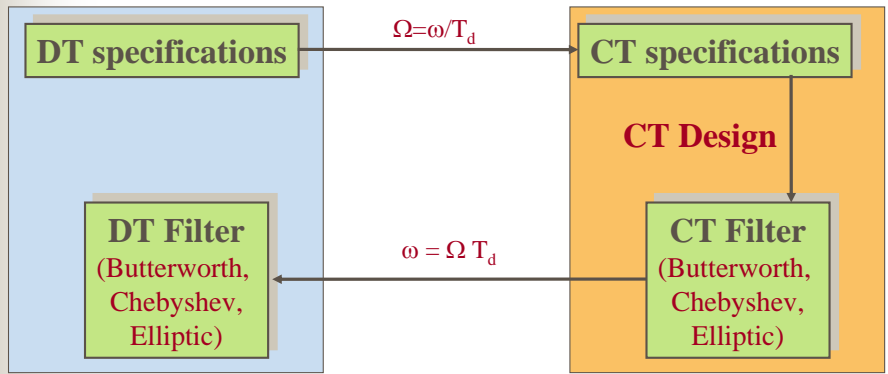
■ Filter Construction

- System function

$$H_c(s)H_c(-s) = \frac{1}{1 + (j\Omega / j\Omega_c)^{2N}}$$

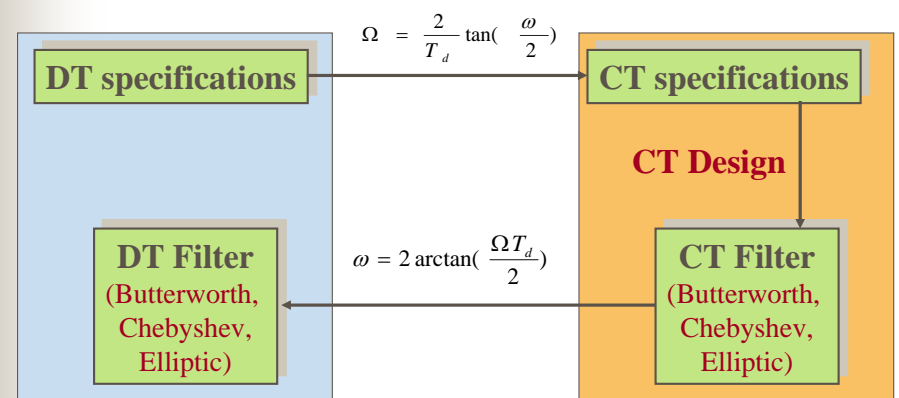
- Roots of denominator polynomial
- Stable and causal system $H_c(s)$

DT IIR Filter by Impulse Invariance



$$H(e^{j\omega}) = H_c(j\frac{\omega}{T_d}) \quad |\omega| \leq \pi \quad H_c(s) = \sum_{k=1}^N \frac{A_k}{s - s_k} \Rightarrow H(z) = \sum_{k=1}^N \frac{T_d A_k}{1 - e^{s_k T_d} z^{-1}}$$

DT IIR Filter Design by Bilinear Transform



$$s = \frac{2}{T_d} \left(\frac{1 - z^{-1}}{1 + z^{-1}} \right), \quad H(z) = H_c \left(\frac{2}{T_d} \left(\frac{1 - z^{-1}}{1 + z^{-1}} \right) \right)$$



How to design some other frequency-selective DT IIR filters besides **lowpass** ones?

Discrete filters

- Lowpass ...
- Highpass...
- Bandpass ...
- Bandstop....

Algebraic transform

A prototype DT lowpass filter

Transform from Lowpass to Another Lowpass

■ Preknowledge

a prototype lowpass filter $H_{lp}(Z)$ with cutoff frequency θ_p

■ Transform

$$Z^{-1} = \frac{z^{-1} - \alpha}{1 - \alpha z^{-1}}$$

Desired cutoff freq.

■ Design Parameter

$$\alpha = \frac{\sin\left(\frac{\theta_p - \omega_p}{2}\right)}{\sin\left(\frac{\theta_p + \omega_p}{2}\right)}$$

■ Wanted lowpass $H(z) = H_{lp}(Z) \Big|_{Z^{-1} = \frac{z^{-1} - \alpha}{1 - \alpha z^{-1}}}$



Prototype filter
 Open matlab:
 >>freqz([0.125],[1 -0.8825])
 Then, Bode plot

Exercise: Design a Highpass Filter (see distributed paper)

Designed filter
 Open matlab:
 >> freqz([0.125 -0.0939],[0.3373 0.1316])
 Then, Bode plot

Transform from Lowpass to a Highpass

- **Preknowledge**
 a prototype lowpass filter $H_{lp}(Z)$ with cutoff frequency θ_p

- **Transform** $Z^{-1} = -\frac{z^{-1} + \alpha}{1 + \alpha z^{-1}}$

- **Design Parameter** $\alpha = -\frac{\cos(\frac{\theta_p + \omega_p}{2})}{\cos(\frac{\theta_p - \omega_p}{2})}$
 (Note: ω_p is circled and labeled "Desired cutoff freq.")

- **Wanted highpass** $H(z) = H_{lp}(Z) \Big|_{Z^{-1} = -\frac{z^{-1} + \alpha}{1 + \alpha z^{-1}}}$

Discrete-Time Linear Phase Filters



Effect of Filtering

- System frequency response:
 $H(e^{j\omega}) = |H(e^{j\omega})| e^{j\angle H(e^{j\omega})}$
- Input and output relationship
 $|Y(e^{j\omega})| = |H(e^{j\omega})| |X(e^{j\omega})|$
 $\angle Y(e^{j\omega}) = \angle H(e^{j\omega}) + \angle X(e^{j\omega})$