EDISP (Filters intro) (English) Digital Signal Processing Digital/Discrete Time/ filters introduction lecture

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DT system characteristics



but we are interested in its characteristics in the frequency domain

$$H(e^{j\theta}) = H(z)|_{z=e^{j\theta}} = \sum_{n=-\infty}^{\infty} h(n)e^{-jn\theta}$$

Filter design



Approximation: find best rational function

$$\frac{b_0 + b_1 z^{-1} + \ldots + b_M z^{-M}}{1 + a_1 z^{-1} + \ldots + a_N z^{-N}} \quad (IIR)$$

or

$$b_0 + b_1 z^{-1} + \ldots + b_M z^{-M}$$
 (FIR)

determine order and coefficients, check stability

Implementation: structure, noise, hardware/software ...

FIR filter design – window method

► Ideal filter:
$$A_0(\theta) = \begin{cases} 1 & \text{for} & |\theta| < \theta_p \\ 0 & \text{for} & \theta_p < |\theta| \le \pi \end{cases}$$
 and zero phase

Impulse response:

$$h_0(n) = \frac{1}{2\pi} \int_{-\pi}^{\pi} H_0(e^{i\theta}) e^{in\theta} d\theta = \frac{\theta_p}{\pi} \frac{\sin n\theta_p}{n\theta_p}$$

is non-causal and infinite!

- Make it finite: $h_P[n] = h_0[n]g[n](g[n] = 0 \text{ for } |n| > P)$
- Shift it to be causal delay by P samples: $h[n] = h_P[n-P]$
- \longrightarrow finally we obtain

$$H(z) = \sum_{n=0}^{2P} h(n) z^{-n} = z^{-P} H_P(z)$$

Implementations

- Transversal (linear convolution)
- ► By IFFT(FFT(x) · FFT(h))
- Lagrange structure: FIR with IIR inside