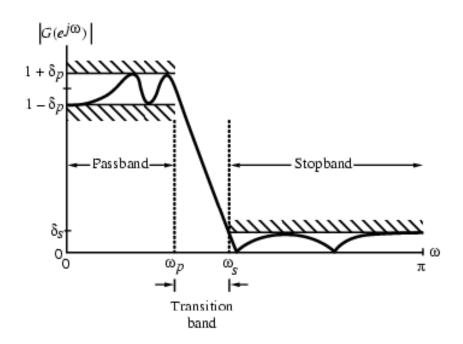
The magnitude response of a digital lowpass filter may be given as indicated below



Filter specification parameters

- ω_p passband edge frequency
- ω_s stopband edge frequency
- δ_p peak ripple value in the passband
- $\delta_{\rm s}$ peak ripple value in the stopband

- Practical specifications are often given in terms of loss function (in dB)
- $G(\omega) = -20\log_{10} \left| G(e^{j\omega}) \right|$
- Peak passband ripple

$$\alpha_p = -20\log_{10}(1 - \delta_p) \text{ dB}$$

Minimum stopband attenuation

$$\alpha_s = -20\log_{10}(\delta_s) dB$$

- In practice, passband edge frequency F_p and stopband edge frequency F_s are specified in Hz
- For digital filter design, normalized bandedge frequencies need to be computed from specifications in Hz using

$$\omega_p = \frac{\Omega_p}{F_T} = \frac{2\pi F_p}{F_T} = 2\pi F_p T$$

$$\omega_s = \frac{\Omega_s}{F_T} = \frac{2\pi F_s}{F_T} = 2\pi F_s T$$

- Example LetkHz
- kHz, kH $F_p = 7$
- kHz, and $F_s = 3$

Then

$$F_T = 25$$

$$\omega_p = \frac{2\pi (7 \times 10^3)}{25 \times 10^3} = 0.56\pi$$

$$\omega_s = \frac{2\pi (3 \times 10^3)}{25 \times 10^3} = 0.24\pi$$

Selection of Filter Type

- The transfer function H(z) meeting the specifications must be a causal transfer function
- For IIR real digital filter the transfer function is a real rational function of

$$H(z) = \frac{p_0 + p_1 z^{-1} + p_2 z^{-2} + z^{-1} + p_M z^{-M}}{d_0 + d_1 z^{-1} + d_2 z^{-2} + \dots + d_N z^{-N}}$$

 H(z) must be stable and of lowest order N or M for reduced computational complexity

Selection of Filter Type

• FIR real digital filter transfer function is a polynomial in z_0 -1 (order N) with real coefficients

$$H(z) = \sum_{n=0}^{N} h[n] z^{-n}$$

- For reduced computational complexity, degree N of H(z) must be as small as possible
- If a linear phase is desired then we must have:

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

(More on this later)