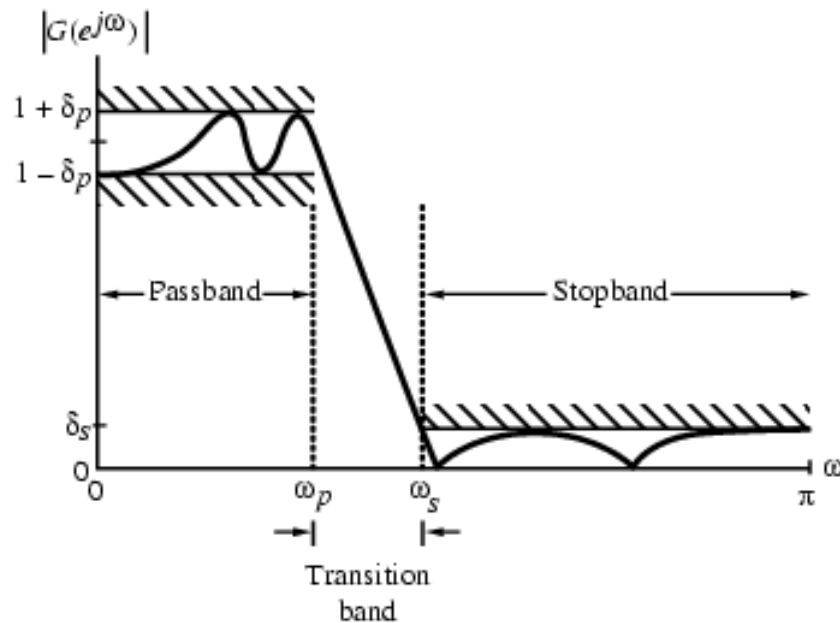


# Digital Filter Specifications

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



# Digital Filter Specifications

Filter specification parameters

- $\omega_p$  - **passband edge frequency**
- $\omega_s$  - **stopband edge frequency**
- $\delta_p$  - **peak ripple value in the passband**
- $\delta_s$  - **peak ripple value in the stopband**

# Digital Filter Specifications

- Practical specifications are often given in terms of **loss function (in dB)**

- $$G(\omega) = -20\log_{10}|G(e^{j\omega})|$$

- **Peak passband ripple**

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

- **Minimum stopband attenuation**

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

# Digital Filter Specifications

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

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

# Digital Filter Specifications

- Example - Let  $F_p = 7$  kHz,  $F_s = 3$  kHz, and  $F_T = 25$  kHz
- Then

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

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

# 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

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

- $H(z)$  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  $z^{-1}$  (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)