Problem 1

We consider a band-limited signal x(t),

$$X(j\Omega) = 0$$
 if $|\Omega| \ge |\Omega_1$.

We consider the impulse response of a filter, h(t), which is also band-limited,

$$H(j\Omega) = 0$$
 if $|\Omega| \ge |\Omega_2$.

The continuous signal x(t) is convolved with the filter h(t),

$$y(t) = h * x(t).$$

The signal y(t) is then sampled with a sampling period T,

$$y[n] = y(nT)$$
.

Finally, we also sample the impulse response h(t) with the sampling period T,

$$h[n] = h(nT)$$
.

We recall the following identity,

$$X(e^{j\omega}) = X_s(j\omega/T),$$

where X_s is the Fourier transform of the analog sampled signal, $x_s(t)$, obtained by sampling x(t) with the period T. You can apply this identity to any signal sampled with the period T.

- 1. Express the Fourier transform of y[n], $Y(e^{j\omega})$, as a function of $X(j\Omega)$ and $H(j\Omega)$.
- 2. We wish to reverse the order of the sampling and filtering. We therefore sample x(t) with the same sampling period, T,

$$x[n] = x(nT).$$

Express the Fourier transform of $x[n], X(e^{j\omega})$, as a function of $X(j\Omega)$

3. We then filter the digital signal x[n] using the digital filter,

$$g[n] = Th[n].$$

Express the Fourier transform of g[n], $G(e^{j\omega})$, as a function of $H(j\Omega)$.

4. Let w[n] be the output of the convolution of x[n] and g[n],

$$w[n] = g * x[n].$$

Express the Fourier transform of w[n], $W(e^{j\omega})$, as a function of $X(j\Omega)$ and $H(j\Omega)$.

5. What is the condition on T for the output of the two systems to be equal,

$$\forall n\in\mathbb{Z},w[n]=y[n].$$

Hint: this is equivalent to proving that $W(e^{j\omega}) = Y(e^{j\omega})$.

Consider the multirate system shown in Fig. 1 with input x[n]. The Fourier transform of x[n] and the frequency response of H_0 and H_1 are shown in Fig. 2. Let $Y_0(e^{j\omega})$ and $Y_1(e^{j\omega})$ be the Fourier transforms of the outputs $y_0[n]$ and $y_1[n]$, respectively.

- 1. Express $Y_0(e^{j\omega})$ and $Y_1(e^{j\omega})$ in terms of $X(e^{j\omega})$.
- 2. Sketch $Y_0(e^{j\omega})$ and $Y_1(e^{j\omega})$.

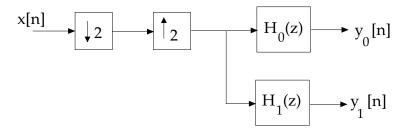
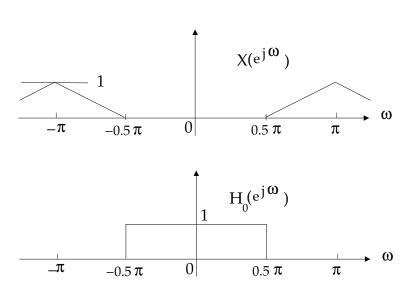


Figure 1:



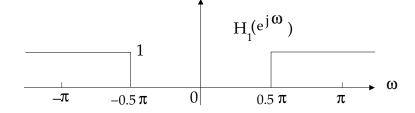


Figure 2:

Problem 3

The continuous time signal $x_c(t)$ is band limited, and has the Fourier transform

$$X_{c}(j\Omega) = \begin{cases} \frac{\pi}{\Omega_{0}} \left[1 + \cos(\frac{\pi\Omega}{\Omega_{0}}) \right] & \text{if } |\Omega| \leq \Omega_{0}, \\ 0 & \text{otherwise.} \end{cases}$$
 (1)

The signal $x_c(t)$ is sampled with a period $T = 2\pi/\Omega_0$.

1. Determine the expression of $x[n] = x_c(nT)$.

Problem 4

We consider a type I generalized linear phase filter h(n) with real coefficients and order 2M. Let $H(\omega)$ be the frequency response of h(n). We assume that H is a low pass filter with the following specifications :

$$\begin{cases} 1 - \delta_p \le H(\omega) \le 1 + \delta_p & \text{if } \omega \in [0, \omega_p), \\ -\delta_s \le H(\omega) \le \delta_s & \text{if } \omega \in [\omega_s, \pi), \end{cases}$$
 (2)

with $\delta_s < 1 - \delta_p$.

1. We define a new filter g(n) as follows

$$g(n) = \begin{cases} h(n) & \text{if } n \neq M, \\ h(M) + \delta_s & \text{if } n = M. \end{cases}$$
 (3)

Prove that $G(\omega)$, the frequency response of g(n), can be written as

$$G(\omega) = P(\omega)e^{-jM\omega},\tag{4}$$

where $P(\omega)$ is real and positive : $P(\omega) \ge 0$.

2. Prove that there exists a minimum phase filter Q(z) such that

$$P(z) = Q(z)Q(1/z). (5)$$

Problem 5

Let H_1 and H_2 be two first order stable all-pass filters. Prove that if

$$H(z) = H_1(z) + H_2(z)$$
(6)

is all-pass then $H_1 = H_2$.