

Nepal Engineering College
Digital Signal Analysis & Processing
Answer of some selected questions

1. A digital communication link carries binary-coded words representing samples of an input signal,

$$x(t) = 3\cos 600\pi t + 2\cos 1800\pi t$$

The link is operated at 10,000 bits/s and each input sample is quantized into 1024 different voltage levels.

- i. What is the sampling frequency & folding frequency?
- ii. What is the Nyquist rate for the signal $x(t)$?
- iii. What are the frequencies in the resulting discrete time signal $x[n]$?
- iv. What is the resolution Δ ?

Solution: ➔

As the link is operated at 10,000 bits/s & each input sample is quantized into 1024 different voltage levels the each sampled value is represented by $\log_2 1024 = 10$ bits/sample

- i. Then maximum sampling frequency, $F_s = \frac{10,000 \text{ bits/sec}}{10 \text{ bits/sample}} = 1000 \text{ samples/sec}$

Folding frequency (is the maximum frequency that can be represented uniquely by sampled signal), $\frac{F_s}{2} = 500 \text{ samples/sec}$.

- ii. $x(t) = 3\cos 600\pi t + 2\cos 1800\pi t$

Here, $F_1 = 300 \text{ Hz}$ and $F_2 = 900 \text{ Hz}$. Thus $F_{\max} = 900 \text{ Hz}$.

The Nyquist rate, $F_N = 2F_{\max} = 1800 \text{ Hz}$.

- iii. For sampling frequency, $F_s = 1000 \text{ Hz}$,

$$\begin{aligned} x[n] &\cong x(nT) = x\left(\frac{n}{F_s}\right) = 3\cos 2\pi \left(\frac{300}{1000}\right)n + 2\cos 2\pi \left(\frac{900}{1000}\right)n \\ &= 3\cos 2\pi \left(\frac{3}{10}\right)n + 2\cos 2\pi \left(\frac{9}{10}\right)n \end{aligned}$$

(Here frequency $f_2 = 9/10$ is greater than $1/2$, so)

$$\begin{aligned} &= 3\cos 2\pi \left(\frac{3}{10}\right)n + 2\cos 2\pi \left(1 - \frac{1}{10}\right)n \\ &= 3\cos 2\pi \left(\frac{3}{10}\right)n + 2\cos 2\pi \left(\frac{1}{10}\right)n \\ \therefore f_1 &= \frac{3}{10} \text{ \& } f_2 = \frac{1}{10} \end{aligned}$$

Here both frequencies f_1 and f_2 lies in the interval $-\frac{1}{2} \leq f \leq \frac{1}{2}$

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iv. ADC resolution = 10 bits

$$\text{Voltage resolution, } \Delta = \frac{x_{\max} - x_{\min}}{L-1} = \frac{5 - (-5)}{1024-1} = 9.76 \text{ mV}.$$

2. Check for linearity, time invariant and stability of following system:

$$y[n] = x[n^2 - 3]u[n] + x[n]u[n]$$

Solution →

For Linearity:

$$\begin{aligned} y_1[n] &= x_1[n^2 - 3]u[n] + x_1[n]u[n] \\ y_2[n] &= x_2[n^2 - 3]u[n] + x_2[n]u[n] \\ \text{let } x_3[n] &= a_1x_1[n] + a_2x_2[n] \\ \therefore y_3[n] &= x_3[n^2 - 3]u[n] + x_3[n]u[n] \\ &= \{a_1x_1[n^2 - 3] + a_2x_2[n^2 - 3]\}u[n] + \{a_1x_1[n] + a_2x_2[n]\}u[n] \\ &= a_1\{x_1[n^2 - 3]u[n] + x_1[n]u[n]\} + a_2\{x_2[n^2 - 3]u[n] + x_2[n]u[n]\} \\ &= a_1y_1[n] + a_2y_2[n] \end{aligned}$$

Since $y_3[n] = a_1y_1[n] + a_2y_2[n]$, the system is linear.

For Time invariant:

$$\text{Response of delayed input, } y(n, k) = x[n^2 - 3 - k]u[n] + x_1[n - k]u[n]$$

$$\text{Delayed response, } y[n - k] = x[(n - k)^2 - 3]u[n - k] + x[n - k]u[n - k]$$

Since $y(n, k) \neq y[n - k]$, the system is time variant.

For Stability:

$$\text{Bounded Input condition: } |x[n]| \leq M_x < \infty$$

For bounded output,

$$\begin{aligned} |y[n]| &= |x[n^2 - 3]u[n] + x[n]u[n]| \\ &\leq |x[n^2 - 3]u[n]| + |x[n]u[n]| \\ &\leq M_x + M_x \leq M_y < \infty \end{aligned}$$

Here, the output is bounded, therefore the system is stable.

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3. The impulse response $h[n]$ of an LTI system is known to be zero, except in the interval $N_0 \leq n \leq N_1$. The input $x[n]$ is known to be zero, except in the interval $N_2 \leq n \leq N_3$. As a result, the output is constrained to be zero, except in some interval $N_4 \leq n \leq N_5$. Determine N_4 and N_5 in terms of N_0, N_1, N_2 , and N_3 .

Solution: \rightarrow

For an LTI system the output is obtained from the convolution of the input with impulse response of the system:

$$y[n] = \sum_{k=-\infty}^{\infty} h[k]x[n-k]$$

Since $h[n] \neq 0$ for $N_0 \leq n \leq N_1$

$$y[n] = \sum_{k=N_0}^{N_1} h[k]x[n-k]$$

The input $x[n] \neq 0$ for $N_2 \leq n \leq N_3$, so

$x[n-k] \neq 0$ for $N_2 \leq (n-k) \leq N_3$

Note the minimum value of $(n-k)$ is N_2 . Thus the lower bound on n , which occurs for $k = N_0$ is

$$N_4 = N_0 + N_2$$

Using similar argument

$$N_5 = N_1 + N_3 \quad N_0 + N_2 \leq n \leq N_1 + N_3$$

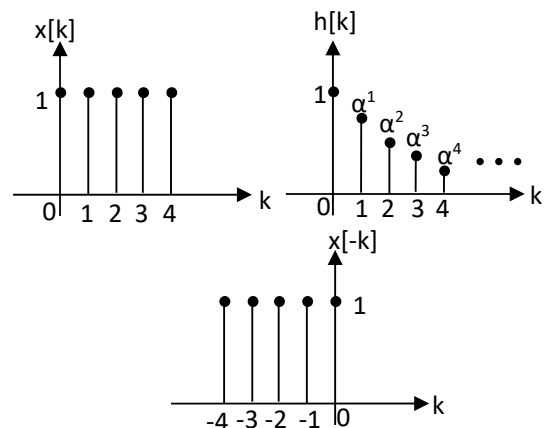
4. Find the response of the system to the input signal $x[n] = u[n] - u[n-5]$. The impulse response of the system is given by $h[n] = \alpha^n u[n]$.

Solution: From figure

$$\begin{aligned} y[n] &= 0 \text{ for } n < 0 \\ y[0] &= h[0]x[-0] = 1 \\ y[1] &= h[1]x[1-1] = 1 + \alpha \\ y[2] &= h[2]x[2-2] = 1 + \alpha + \alpha^2 \\ y[3] &= h[3]x[3-3] = 1 + \alpha + \alpha^2 + \alpha^3 \\ y[4] &= h[4]x[4-4] = 1 + \alpha + \alpha^2 + \alpha^3 + \alpha^4 \\ y[5] &= h[5]x[5-5] = \alpha + \alpha^2 + \alpha^3 + \alpha^4 + \alpha^5 \\ y[6] &= h[6]x[6-6] = \alpha^2 + \alpha^3 + \alpha^4 + \alpha^5 + \alpha^6 \end{aligned}$$

And so on..

Therefore,



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$$y[n] = \begin{cases} 0, & \text{for } n < 0 \\ \sum_{k=0}^n \alpha^k, & 0 \leq n \leq 4 \\ \sum_{k=n-4}^n \alpha^k, & \text{for } n > 4 \end{cases}$$

5. Find z-transform of $x[n] = a^n(\cos\omega_0 n)u[n]$.

Solution →

$$x[n] = a^n(\cos\omega_0 n)u[n] = \frac{1}{2}a^n e^{j\omega_0 n}u[n] + \frac{1}{2}a^n e^{-j\omega_0 n}u[n]$$

We can write the above expression as $x[n] = v[n] + v^*[n]$

$$\text{where } v[n] = \frac{1}{2}\alpha^n u[n] \text{ with } \alpha = ae^{j\omega_0}$$

The z-transform $V(z)$ of $v[n]$ by linearity and scaling in z-domain properties:

$$V(z) = \frac{1}{2} \cdot \frac{1}{1 - \alpha z^{-1}} = \frac{1}{2} \cdot \frac{1}{1 - ae^{j\omega_0} z^{-1}}, |z| > |\alpha| = a$$

From conjugate of complex sequence property,

$$V^*(z^*) = \frac{1}{2} \cdot \frac{1}{1 - \alpha^* z^{-1}} = \frac{1}{2} \cdot \frac{1}{1 - ae^{-j\omega_0} z^{-1}}, |z| > |\alpha| = a$$

By linearity property of the z-transform, we obtain

$$\begin{aligned} X(z) &= V(z) + V^*(z^*) \\ &= \frac{1}{2} \left(\frac{1}{1 - ae^{j\omega_0} z^{-1}} + \frac{1}{1 - ae^{-j\omega_0} z^{-1}} \right) = \frac{1 - (a\cos\omega_0)z^{-1}}{1 - (2a\cos\omega_0)z^{-1} + a^2 z^{-2}} \quad |z| > a \end{aligned}$$

6. Given the system $y[n] = 3y[n-1] - 0.1y[n-2] + x[n]$

- a. Find the system function, $H(z)$.**
- b. Is the system IIR or FIR?**
- c. Plot the pole-zero diagram.**
- d. Plot the magnitude response. [PU Fall 2012 5(a)]**

Solution: →

- a. Taking z-transform of both sides, we get

$$Y(z) = 3z^{-1}Y(z) - 0.1z^{-2}Y(z) + X(z)$$

$$\Rightarrow Y(z)\{1 - 3z^{-1} + 0.1z^{-2}\} = X(z)$$

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$$\therefore H(z) = \frac{Y(z)}{X(z)} = \frac{1}{1 - 3z^{-1} + 0.1z^{-2}}$$

This is the required System function of given system.

- b. As the system contains poles the given system is IIR. (If the system contain poles then system is IIR)
c. Also,

$$H(z) = \frac{1}{1 - 3z^{-1} + 0.1z^{-2}} = \frac{1}{(1 - 2.9663z^{-1})(1 - 0.0337z^{-1})}$$

The system contains poles at $p_1 = 2.966$ and $p_2 = 0.034$, and zeros are at $z_1 = z_2 = 0$, the pole-zero diagram is shown below:

- d. For the frequency response, we will take $z = e^{j\omega}$

$$\therefore H(e^{j\omega}) = \frac{1}{(1 - 2.9663e^{-j\omega})(1 - 0.0337e^{-j\omega})}$$

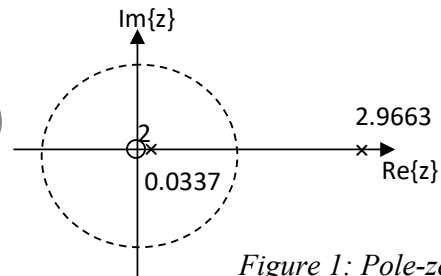


Figure 1: Pole-zero plot

Magnitude is given by

$$|H(e^{j\omega})| = \left| \frac{1}{(1 - 2.9663e^{-j\omega})(1 - 0.0337e^{-j\omega})} \right| = \frac{1}{|1 - 2.9663e^{-j\omega}| |1 - 0.0337e^{-j\omega}|}$$

$$= \frac{1}{\sqrt{(1 - 2.9663\cos\omega)^2 + (2.9663\sin\omega)^2} \cdot \sqrt{(1 - 0.0337\cos\omega)^2 + (0.0337\sin\omega)^2}}$$

For the magnitude response, we can

find the ω Vs. $|H(e^{j\omega})|$ as,

ω	$ H(e^{j\omega}) $
0	0.5263
$\pi/6$	0.4770
$2\pi/6$	0.3890
$3\pi/6$	0.3193
$4\pi/6$	0.2751
$5\pi/6$	0.2514
π	0.2493

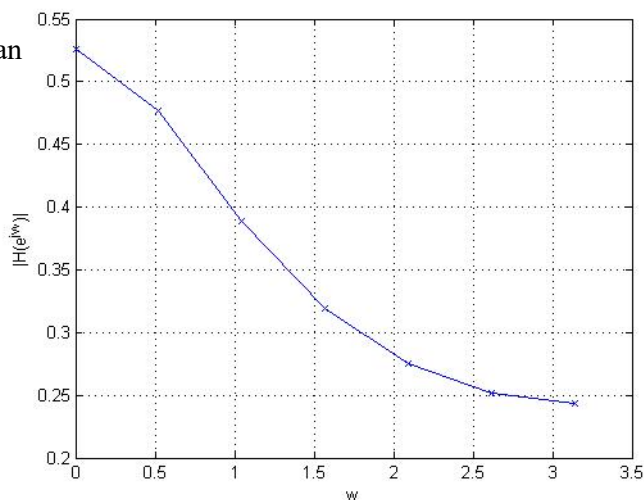


Figure 2: Magnitude Response

7. Determine the inverse z-transform of

$$H(z) = \frac{1 + 2z^{-1} + z^{-2}}{1 + 4z^{-1} + 4z^{-2}}$$

For all possible ROCs. [PU 2009 Spring, 3(a)].

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Solution: →

We have,

$$\begin{aligned} H(z) &= \frac{1 + 2z^{-1} + z^{-2}}{1 + 4z^{-1} + 4z^{-2}} = \frac{1 + 2z^{-1} + z^{-2}}{(1 + 2z^{-1})^2} \\ &= \frac{1}{(1 + 2z^{-1})^2} + \frac{2z^{-1}}{(1 + 2z^{-1})^2} + \frac{z^{-2}}{(1 + 2z^{-1})^2} \\ &= -0.5z \times \frac{-2z^{-1}}{(1 - (-2)z^{-1})^2} + (-1) \times \frac{-2z^{-1}}{(1 - (-2)z^{-1})^2} + (-0.5)z^{-1} \times \frac{-2z^{-1}}{(1 - (-2)z^{-1})^2} \end{aligned}$$

The system have poles at $p_1 = -2$ and $p_2 = -2$. Therefore the possible ROCs are $|z| > 2$ and $|z| < 2$.

a. For ROC $|z| > 2$ (Causal Signal)

$$\therefore h[n] = -0.5(n+1)(-2)^{n+1}u[n+1] - n(-2)^n u[n] - 0.5(n-1)(-2)^{n-1}u[n-1]$$

b. For ROC $|z| < 2$ (Anticausal signal)

$$\begin{aligned} \therefore h[n] &= 0.5(n+1)(-2)^{n+1}u[-n-2] + n(-2)^n u[-n-1] \\ &\quad + 0.5(n-1)(-2)^{n-1}u[-n] \end{aligned}$$

Note: We have used shifting property of z-transform and following z-transform table:

$$\begin{aligned} na^n u[n] &\xleftrightarrow{z} \frac{az^{-1}}{(1 - az^{-1})^2}, \quad |z| > |a| \\ -na^n u[-n-1] &\xleftrightarrow{z} \frac{az^{-1}}{(1 - az^{-1})^2}, \quad |z| < |a| \end{aligned}$$

After using shifting property,

$$\begin{aligned} (n-k)a^{n-k}u[n-k] &\xleftrightarrow{z} z^{-k} \times \frac{az^{-1}}{(1 - az^{-1})^2}, \quad |z| > |a| \\ -(n-k)a^{n-k}u[-(n-k)-1] &\xleftrightarrow{z} z^{-k} \times \frac{az^{-1}}{(1 - az^{-1})^2}, \quad |z| < |a| \end{aligned}$$

8. If $y[n] = \{1, 1, 2, -1, 3\}$ is output and impulse response is $h[n] = \{1, 2, 3\}$ find discrete-time signal $x[n]$.

Solution: →

Taking z-transform,

$$\begin{aligned} Y(z) &= 1 + z^{-1} + 2z^{-2} - z^{-3} + 3z^{-4} \\ H(z) &= 1 + 2z^{-1} + 3z^{-2} \\ \therefore X(z) &= \frac{Y(z)}{H(z)} = \frac{1 + z^{-1} + 2z^{-2} - z^{-3} + 3z^{-4}}{1 + 2z^{-1} + 3z^{-2}} \end{aligned}$$

By division method:

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$$\begin{array}{r} 1 + 2z^{-1} + 3z^{-2} \mid 1 + z^{-1} + 2z^{-2} - z^{-3} + 3z^{-4} \mid 1 - z^{-1} + z^{-2} \\ \underline{1 + 2z^{-1} + 3z^{-2}} \\ -z^{-1} - z^{-2} - z^{-3} + 3z^{-4} \\ \underline{-z^{-1} - 2z^{-2} - 3z^{-3}} \\ z^{-2} + 2z^{-3} + 3z^{-4} \\ \underline{z^{-2} + 2z^{-3} + 3z^{-4}} \\ 0 \end{array}$$

Therefore $X(z)$ is given by,

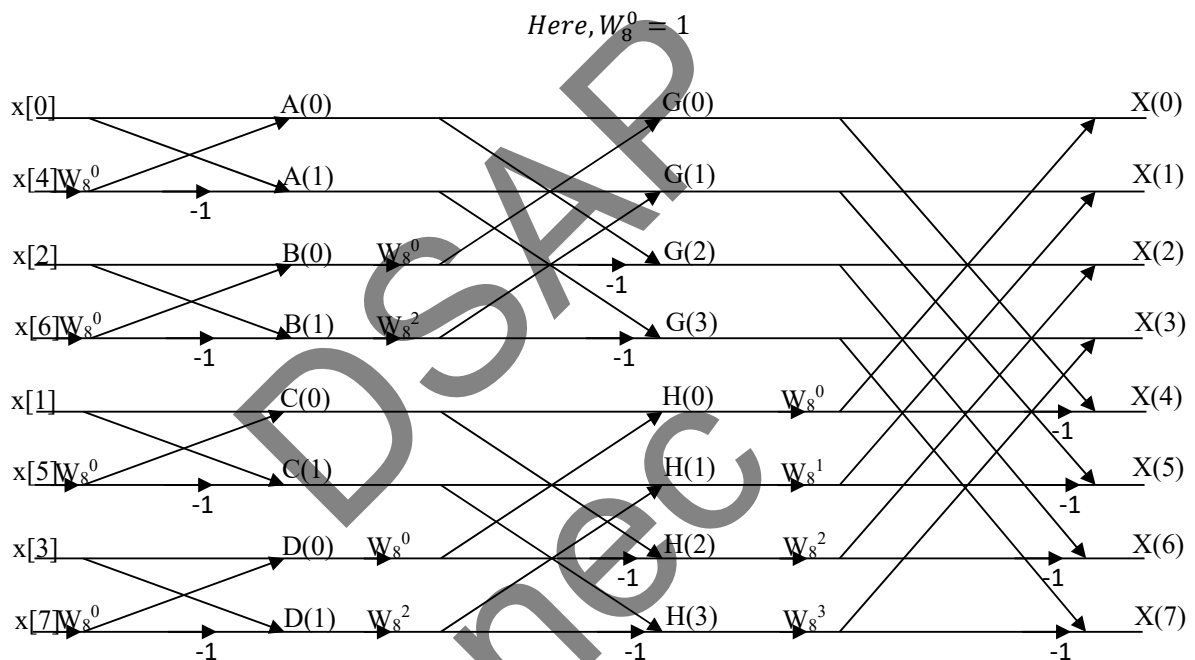
$$X(z) = 1 - z^{-1} + z^{-2}$$

Taking inverse z-transform, we get,

$$x[n] = \{1, -1, 1\}.$$

9. Find fft of the signal $x[n] = \{1, 2, 5, 4, -3, 0, 4, 1\}$.

Solution: →



$$W_8^1 = e^{-j\frac{2\pi}{8}} = \cos \frac{2\pi}{8} - j \sin \frac{2\pi}{8} = 0.707 - j0.707$$

$$W_8^2 = e^{-j(\frac{2\pi}{8}) \times 2} = \cos \frac{\pi}{2} - j \sin \frac{\pi}{2} = -j$$

$$W_8^3 = e^{-j(\frac{2\pi}{8}) \times 3} = \cos \frac{3\pi}{4} - j \sin \frac{3\pi}{4} = 0.707 - j0.707$$

Output of first Stage:

$$A(0) = x[0] + W_8^0 x[4] = 1 + 1 \times -3 = -2, \quad A(1) = x[0] - W_8^0 x[4] = 1 - 1 \times -3 = 4$$

$$B(0) = x[2] + W_8^0 x[6] = 5 + 1 \times 4 = 9, \quad B(1) = x[2] - W_8^0 x[6] = 5 - 1 \times 4 = 1$$

$$C(0) = x[1] + W_8^0 x[5] = 2 + 1 \times 0 = 2, \quad C(1) = x[1] - W_8^0 x[5] = 2 - 1 \times 0 = 2$$

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$$D(0) = x[3] + W_8^0 \cdot x[7] = 4 + 1 \times 1 = 5, \quad D(1) = x[3] - W_8^0 \cdot x[7] = 4 - 1 \times 1 = 3$$

Output of second stage:

$$G(0) = A(0) + W_8^0 \cdot B(0) = -2 + 1 \times 9 = 7, \quad H(0) = C(0) + W_8^0 \cdot D(0) = 2 + 1 \times 5 = 7$$

$$G(1) = A(1) + W_8^2 \cdot B(1) = 4 + (-j) \times 1 = 4 - j, \\ H(1) = C(1) + W_8^2 \cdot D(1) = 2 + (-j) \times 3 = 2 - j3$$

$$G(2) = A(0) - W_8^0 \cdot B(0) = -2 - 1 \times 9 = -11, \quad H(2) = A(0) - W_8^0 \cdot B(0) = 2 - 1 \times 5 = -3$$

$$G(3) = A(1) - W_8^2 \cdot B(1) = 4 - (-j) \times 1 = 4 + j, \\ H(3) = C(1) - W_8^2 \cdot D(1) = 2 - (-j) \times 3 = 2 + j3$$

Output of third stage:

$$X(0) = G(0) + W_8^0 \cdot H(0) = 7 + 1 \times 7 = 14$$

$$X(1) = G(1) + W_8^1 \cdot H(1) = (4 - j) + (0.707 - j0.707) \times (2 - j3) = 3.293 - j4.535$$

$$X(2) = G(2) + W_8^2 \cdot H(2) = -11 + (-j) \times (-3) = -11 + j3$$

$$X(3) = G(3) + W_8^2 \cdot H(3) = (4 + j) + (-0.707 - j0.707) \times (2 + j3) = 4.707 - j2.535$$

$$X(4) = G(0) - W_8^0 \cdot H(0) = 7 - 1 \times 7 = 0$$

$$X(5) = G(1) - W_8^1 \cdot H(1) = (4 - j) - (0.707 - j0.707) \times (2 - j3) = 4.707 + j2.535$$

$$X(6) = G(2) - W_8^2 \cdot H(2) = -11 - (-j) \times (-3) = -11 - j3$$

$$X(7) = G(3) - W_8^2 \cdot H(3) = (4 + j) - (-0.707 - j0.707) \times (2 + j3) = 3.293 + j4.535$$

$$\therefore X(k) = \{14, 3.293 - j4.535, -11 + j3, 4.707 - j2.535, 0, 4.707 + j2.535, -11 - j3, 3.293 + j4.535\}$$

10. Draw the lattice structure of the following system:

$$H(z) = \frac{1 + 3z^{-1} + 3z^{-2} + z^{-3}}{2 - 1.8z^{-1} + 1.28z^{-2} - 1.152z^{-3}}. \text{ Is this system stable?}$$

Solution: →

$$\text{Given, } H(z) = \frac{C_3(z)}{2A_3(z)} = \frac{1 + 3z^{-1} + 3z^{-2} + z^{-3}}{2(1 - 0.9z^{-1} + 0.64z^{-2} - 0.576z^{-3})}$$

Note: We have to make $\alpha_m(0) = 1$ for mathematical convenience.

$$\text{First consider, } A_3(z) = 1 - 0.9z^{-1} + 0.64z^{-2} - 0.576z^{-3}$$

$$B_3(z) = -0.576 + 0.64z^{-1} - 0.9z^{-2} + z^{-3}. \text{ (reverse polynomial of } A_3(z))$$

$$\text{We know, } k_m = \alpha_m(m), \text{ therefore } k_3 = -0.576$$

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$$\text{Also, } A_{m-1}(z) = \frac{A_m(z) - k_m B_m(z)}{1 - k_m^2}$$

$$\begin{aligned} \text{therefore, } A_2(z) &= \frac{A_3(z) - k_3 B_3(z)}{(1 - k_3^2)} \\ &= \frac{1 - 0.9z^{-1} + 0.64z^{-2} - 0.576z^{-3} - 0.576(-0.576 + 0.64z^{-1} - 0.9z^{-2} + z^{-3})}{1 - 0.576} \\ &= 1 - 0.795z^{-1} + 0.182z^{-2} \\ \therefore k_2 &= 0.182 \end{aligned}$$

Similarly,

$$\begin{aligned} A_1(z) &= \frac{A_2(z) - k_2 B_2(z)}{(1 - k_2^2)} = 1 - 0.672z^{-1} \\ \therefore k_2 &= 0.182 \end{aligned}$$

Now consider the numerator part $C_3(z)$

$$C_3(z) = c_3(0) + c_3(1)z^{-1} + c_3(2)z^{-2} + c_3(3)z^{-3} = 1 + 3z^{-1} + 3z^{-2} + z^{-3}$$

$$\text{We know, } v_m = c_m(m). \text{ Therefore } v_3 = c_3(3) = 1$$

$$\text{Now we have, } C_{m-1}(z) = C_m(z) - v_m B_m(z)$$

$$\therefore C_2(z) = C_3(z) - v_3 B_3(z) \quad (\text{Note: } B_3(z) \text{ is from lattice part})$$

$$\begin{aligned} C_3(z) &= 1 + 3z^{-1} + 3z^{-2} + z^{-3} - 1(-0.576 + 0.64z^{-1} - 0.9z^{-2} + z^{-3}) \\ &= 1.576 + 2.36z^{-1} + 3.9z^{-2} \\ \therefore v_2 &= c_2(2) = 3.9 \end{aligned}$$

$$\text{Similarly, } C_1(z) = C_2(z) - v_2 B_2(z) = 0.866 + 5.46z^{-1} \Rightarrow v_1 = c_1(1) = 5.46$$

$$\text{And, } C_0(z) = C_1(z) - v_1 B_1(z) = 4.535 \Rightarrow v_0 = c_0(0) = C_0(z) = 4.535$$

For Stability, $|k_m| < 1$. Therefore the system is stable.

The lattice ladder structure is shown below:

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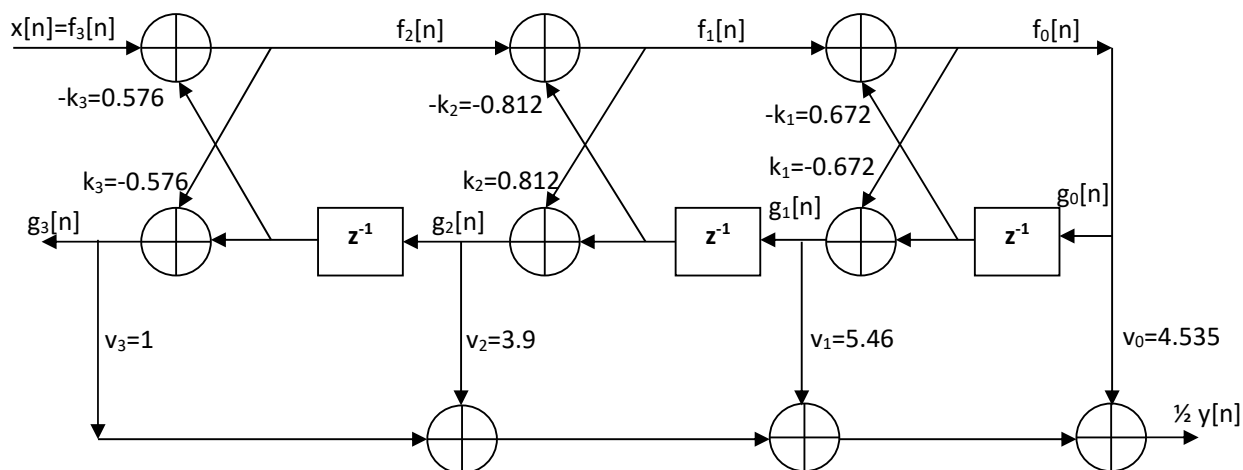


Fig: Lattice-ladder structure of given question.

11. Design an FIR filter with 9 coefficients for the following specifications:

- Passband edge frequency = 0.5 KHz
- Sampling frequency = 2 KHz
- Use suitable window in design.

Solution:

Given,

$$M = 9$$

$$F_c = 0.5 \text{ KHz}$$

$$F_s = 2 \text{ KHz}$$

We know,

$$\omega_c = 2\pi f = 2\pi F_c / F_s = 0.5\pi$$

The desired frequency response is given by

$$H_d(\omega) = \begin{cases} e^{-j\omega\tau} & \text{for } |\omega| \leq \omega_c \\ 0 & \text{elsewhere} \end{cases}$$

Taking inverse discrete-time Fourier transform to obtain desired unit sample response, we get,

$$h_d(n) = \frac{1}{2\pi} \int_{-\pi}^{\pi} H_d(\omega) e^{j\omega n} d\omega = \frac{1}{2\pi} \int_{-0.5\pi}^{0.5\pi} e^{-j\omega\tau} e^{j\omega n} d\omega$$

Solving above equation we get,

$$h_d(n) = \begin{cases} \frac{\sin 0.5\pi(n - \tau)}{\pi(n - \tau)}, & \text{for } n \neq \tau \\ \frac{1}{2}, & \text{for } n = \tau \end{cases}$$

We know for symmetric FIR filter, the conditions are

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$$h[n] = h[M - 1 - n] \text{ \& } \tau = \frac{M-1}{2}$$

for $M = 9$, the value of $\tau = 4$

$$\therefore h_d(n) = \begin{cases} \frac{\sin 0.5\pi(n-4)}{\pi(n-4)}, & \text{for } n \neq 4 \\ \frac{1}{2}, & \text{for } n = 4 \end{cases}$$

As minimum stopband attenuation condition is not given, We can take any window for the design. So, taking Hanning window for our design,

Then the required unit sample is given by,

$$h[n] = h_d[n] \cdot w[n]$$

Taking Hanning window in design, the required unit sample of given filter is given in table below:

N	$h_d[n]$	$w[n] = 0.5 \left(1 - \cos \frac{2\pi n}{M-1} \right)$	$h[n] = h_d[n] \cdot w[n]$
0	0	0	0
1	-0.106	0.146	0.015
2	0	0.5	0
3	0.318	0.854	0.269
4	0.5	1	0.5
5	0.318	0.854	0.269
6	0	0.5	0
7	-0.106	0.146	0.015
8	0	0	0

12. Design an FIR linear phase filter using Kaiser window to meet the following specifications.

$$0.99 \leq |H(e^{jw})| \leq 1.01, \text{ for } 0 \leq |w| \leq 0.19\pi$$

$$|H(e^{jw})| \leq 0.01, \text{ for } 0.21\pi \leq |w| \leq \pi$$

Solution:

Step I: Given data

The given specifications may be rewritten as under

$$1-0.01 \leq |H(e^{jw})| \leq 1+0.01, \text{ for } 0 \leq \omega \leq 0.19\pi$$

$$|H(e^{jw})| \leq 0.01, \text{ for } 0.21\pi \leq \omega \leq \pi$$

On comparing the above given specifications with magnitude response of filter,

$$\delta_1=0.01 \quad \delta_2=0.01$$

$$\omega_p = 0.19\pi \quad \omega_s = 0.21\pi$$

$$\Delta\omega = \omega_s - \omega_p = 0.02\pi$$

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$\delta = \text{minimum of } \delta_1 \text{ and } \delta_2 = 0.01$

Now, $A = -20\log_{10}\delta = 40$

Step 2: To determine cutoff frequency ω_c

Cutoff frequency is given by

$$\omega_c = (\omega_p + \omega_s)/2 = 0.2\pi$$

Step 3: To obtain β and M

Here $A = 40$, which lies in the range of 21 to 51

Hence, β can be obtained as,

$$\beta = 0.5842(A-21)^{0.4} + 0.07886(A-21) = 3.395$$

and $M = \frac{A-8}{2.2854\omega} = 222.88 \cong 223$

Step 4: To obtain expression of Kaiser window

The window will be defined as under

$$w[n] = \frac{I_0\left[\beta \left(\sqrt{1 - \left(\frac{n - \frac{M}{2}}{\frac{M}{2}}\right)^2}\right)\right]}{I_0(\beta)} \text{ for } 0 \leq n \leq M$$

The function $I_0()$ in the above expression may be calculated with the help of following equation

$$I_0(x) = 1 + \sum_{r=1}^{\infty} \left[\frac{(x/2)^r}{r!} \right]^2$$

Since, it is an infinite series, maximum possible terms must be taken to reduce the error.

The ideal desired frequency response is given by

$$H_d(\omega) = \begin{cases} e^{-j\omega\tau\left(\frac{M-1}{2}\right)} & \text{for } |\omega| \leq \omega_c \\ 0 & \text{elsewhere} \end{cases}$$

Now, $h_d(n)$ may be obtained by taking inverse Fourier transform of above equation, i.e

$$h_d(n) = \frac{\sin\left[\omega_c \left(n - \frac{M}{2}\right)\right]}{\pi \left(n - \frac{M}{2}\right)}, \text{ for } 0 \leq n \leq M,$$

Substituting for $\omega_c = 0.2\pi$ and $M = 223$,

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$$h_d(n) = \frac{\sin[0.2\pi(n - 111.5)]}{\pi(n - 111.5)}, \text{ for } 0 \leq n \leq 223,$$

The unit sample response of FIR filter may be obtained by windowing, i.e.

$$h[n] = h_d[n]w[n]$$

$$h(n) = \frac{\sin[0.2\pi(n - 111.5)]}{\pi(n - 111.5)} \frac{I_0 \left[3.395 \left(\sqrt{1 - \left(\frac{n - 111.5}{111.5} \right)^2} \right) \right]}{I_0(3.395)}, \text{ for } 0 \leq n \leq 223,$$

This is the required relationship for unit sample response FIR filter using Kaiser Window.

13. Design a lowpass Butterworth filter to meet the following specifications.

Passband gain = 0.89

Passband frequency edge = 30 Hz

Attenuation = 0.20

Stopband edge = 75 Hz

(Use impulse invariance and Bilinear Transformation method to convert the filter from analog to digital).

Solution:→

We have $\Omega_p = 2\pi \times 30 = 60\pi \text{ rad/s}$ ($F_p = 30 \text{ Hz}$)

$$\Omega_s = 2\pi \times 75 = 150\pi \text{ rad/s} \quad (F_s = 75 \text{ Hz})$$

$$\alpha_{\max} = -20 \log_{10}(0.89) = 1.0122 \text{ dB}$$

$$\alpha_{\min} = -20 \log_{10}(0.20) = 13.9794 \text{ dB}$$

$$\varepsilon^2 = 10^{0.1\alpha_{\max}} - 1 = 10^{0.10122} - 1 = 0.2625 \Rightarrow \varepsilon = 0.5123$$

$$\delta^2 = 10^{0.1\alpha_{\min}} - 1 = 10^{1.39794} - 1 = 24 \Rightarrow \delta = 4.899$$

$$\therefore N = \frac{\log_{10} \left(\frac{\delta}{\varepsilon} \right)}{\log_{10} \left(\frac{\Omega_s}{\Omega_p} \right)} = \frac{\log_{10} \left(\frac{4.899}{0.5123} \right)}{\log_{10} \left(\frac{150\pi}{60\pi} \right)} = \frac{0.9806}{0.3979} = 2.4644$$

Since the number of poles must be an integer, we round up to $N = 3$.

Matching the frequency response exactly at stopband produces,

$$\Omega_c = \frac{\Omega_s}{\delta^{\frac{1}{N}}} = \frac{150\pi}{4.899^{\frac{1}{3}}} = 277.4632$$

[Note: If we were instead to match the frequency response at passband, we would obtain

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$$\Omega_c = \frac{\Omega_p}{\frac{1}{\varepsilon N}} = \frac{60\pi}{0.5123^{\frac{1}{3}}} = 235.5734$$

In principle, any value of the critical frequency that satisfies $\frac{\Omega_p}{\frac{1}{\varepsilon N}} \leq \Omega_c \leq \frac{\Omega_s}{\frac{1}{\delta N}}$ would be valid.]

For $N = 3$ and $\Omega_c = 1$ (Normalized Transfer function)

$$H(s) = \frac{1}{(s+1)(s^2+s+1)}$$

(This can be obtained from formula as well as table given)

This transfer function is normalized for $\Omega_c = 1$. However $\Omega_c = 277.46$, we need to denormalize $H(s)$ by $\Omega_c = 277.46$ rad/s

$$H(s)|_{s=s/\Omega_c} = \frac{\Omega_c^3}{(s+\Omega_c)(s^2+\Omega_c s+\Omega_c^2)} = \frac{2.136 \times 10^7}{(s+277.46)(s^2+277.46s+76984)}$$

i. Using Impulse invariance method:

$$H(s) = \frac{2.136 \times 10^7}{(s+277.46)(s^2+277.46s+76984)} = \frac{A}{s+277.46} + \frac{Bs+c}{s^2+277.46s+76984}$$

Solving we get, $A = 277.46$, $B = -277.46$ and $c = 0$.

$$\begin{aligned} \therefore H(s) &= \frac{277.46}{s+277.46} - \frac{277.46s}{s^2+277.46s+76984} \\ &= \frac{277.46}{s+277.46} - \frac{277.46s}{s^2+2 \times s \times 138.73 + 138.73^2 + 230.29^2} \\ &= \frac{277.46}{s+277.46} - \frac{277.46(s+138.73) - 38492}{(s+138.73)^2 + 230.29^2} \\ &= \frac{277.46}{s+277.46} - \frac{277.46(s+138.73)}{(s+138.73)^2 + 230.29^2} + \frac{\frac{38492}{230.29} * 230.29}{(s+138.73)^2 + 230.29^2} \end{aligned}$$

Using mapping method:

$$\begin{aligned} H(z) &= \frac{277.46}{1 - e^{-277.46T} z^{-1}} - 277.46 \times \frac{1 - e^{-138.73T} \cos(230.29T) z^{-1}}{1 - 2e^{-138.73T} \cos(230.29T) z^{-1} + e^{-277.46T} z^{-2}} \\ &\quad + 167.14 \times \frac{e^{-138.73T} \sin(230.29T) z^{-1}}{1 - 2e^{-138.73T} \cos(230.29T) z^{-1} + e^{-277.46T} z^{-2}} \end{aligned}$$

Analog Domain $H(s)$	Digital Domain $H(z)$
$\frac{1}{s - p_k}$	$\frac{1}{1 - e^{p_k T} z^{-1}}$

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$\frac{s+a}{(s+a)^2+b^2}$	$\frac{1-e^{-aT}\cos(bT)z^{-1}}{1-2e^{-aT}\cos(bT)z^{-1}+e^{-2aT}z^{-2}}$
$\frac{b}{(s+a)^2+b^2}$	$\frac{e^{-aT}\sin(bT)z^{-1}}{1-2e^{-aT}\cos(bT)z^{-1}+e^{-2aT}z^{-2}}$

ii. Using Bilinear Transformation Method:

$$s = \frac{2}{T} \left(\frac{1-z^{-1}}{1+z^{-1}} \right).$$

$$H(s) = \frac{2.136 \times 10^7}{(s+277.46)(s^2+277.46s+76984)}$$

$$\therefore H(z) = \frac{2.136 \times 10^7}{\left(\frac{2}{T} \left(\frac{1-z^{-1}}{1+z^{-1}} \right) + 277.46 \right) \left(\left(\frac{2}{T} \left(\frac{1-z^{-1}}{1+z^{-1}} \right) \right)^2 + 277.46 \frac{2}{T} \left(\frac{1-z^{-1}}{1+z^{-1}} \right) + 76984 \right)}$$

Butterworth Polynomials in Factorized form	
Order(n)	$H(s) = H(s) = \frac{1}{A(s)}, \text{Where } A(s)$
1	$(s+1)$
2	$(s^2+1.4142s+1)$
3	$(s+1)(s^2+s+1)$
4	$(s^2+1.8478s+1)(s^2+0.7654s+1)$
5	$(s+1)(s^2+0.6180s+1)(s^2+1.6180s+1)$
6	$(s^2+1.414s+1)(s^2+0.5176s+1)(s^2+1.9319s+1)$
7	$(s+1)(s^2+0.445s+1)(s^2+1.247s+1)(s^2+1.802s+1)$

14. Design a digital lowpass Butterworth filter to meet the following specifications.

- Passband cutoff frequency: $\omega_p = 0.15\pi$
- Stopband cutoff frequency: $\omega_s = 0.35\pi$
- Passband ripple: $-3dB \leq |H(e^{j\omega})| \leq 0dB, |\omega| \leq \omega_p$
- Stopband ripple: $|H(e^{j\omega})| \leq -20dB, \omega_s \leq |\omega| \leq \pi$

Use impulse invariance method.

Solution:

Given,

$$\alpha_{max} = 3dB, \alpha_{min} = 20dB, \omega_p = 0.15\pi, \omega_s = 0.35\pi$$

$$\varepsilon^2 = 10^{0.1\alpha_{max}} - 1 = 10^{0.3} - 1 = 0.995 \Rightarrow \varepsilon = 0.998$$

$$\delta^2 = 10^{0.1\alpha_{min}} - 1 = 10^2 - 1 = 99 \Rightarrow \delta = 9.95$$

Using the impulse invariance design procedure, we have noted that the relation between frequency in the continuous-time and discrete-time domains is $\omega = \Omega T$, where T is merely a design parameter. Leaving T as an arbitrary constant for now, we obtain

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$$\Omega_p = \frac{\omega_p}{T} = \frac{0.47124}{T}, \text{ and } \Omega_s = \frac{\omega_s}{T} = \frac{1.0996}{T}$$

$$\therefore N = \frac{\log_{10}\left(\frac{\delta}{\epsilon}\right)}{\log_{10}\left(\frac{\Omega_s}{\Omega_p}\right)} = \frac{\log_{10}\left(\frac{9.95}{0.998}\right)}{\log_{10}\left(\frac{1.0996}{0.47124}\right)} = \frac{0.998}{0.369} = 2.712$$

Since the number of poles must be an integer, we round up to $N = 3$.

Matching the frequency response exactly at passband produces,

$$\Omega_c = \frac{\Omega_p}{\frac{1}{\epsilon^{\frac{1}{N}}}} = \frac{0.4716}{T}$$

For $N = 3$ and $\Omega_c = 1$ (Normalized Transfer function)

$$H(s) = \frac{1}{(s+1)(s^2+s+1)}$$

(This can be obtained from formula as well as table given)

This transfer function is normalized for $\Omega_c = 1$. However $\Omega_c = 0.4716/T$, we need to denormalise $H(s)$ by $\Omega_c = 0.4716/T$

$$H(s) \big|_{s=s/\Omega_c} = \frac{\Omega_c^3}{(s + \Omega_c)(s^2 + \Omega_c s + \Omega_c^2)} = \frac{\left(\frac{0.4716}{T}\right)^3}{\left(s + \frac{0.4716}{T}\right)\left(s^2 + \frac{0.4716}{T}s + \left(\frac{0.4716}{T}\right)^2\right)}$$

Using partial fractions we can rewrite the system function of the continuous-time prototype filter as

$$H(s) = \sum_{k=1}^N \frac{A_k}{s - s_k},$$

where the parameters A_k are the continuous time residues of the poles s_k .

Using the MATLAB routine *residue* we find the residues for the three poles, producing the transfer function

$$H(s) = \frac{0.4719}{s + \frac{0.4716}{T}} + \frac{0.236 - 0.136j}{s - \frac{-0.236 + 0.408j}{T}} + \frac{-0.236 + 0.136j}{s - \frac{-0.236 - 0.408j}{T}}$$

The corresponding discrete-time filter has the transfer function

$$H(z) = \sum_{k=1}^N \frac{A_k}{1 - e^{s_k T} z^{-1}}$$

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Note that the poles of the continuous-time filter, s_k are all of the form of a (generally complex) constant divided by the parameter T . Since s_k is multiplied by T wherever it appears in the equation for $H(z)$ above, the specific value chosen for T has no effect at all on the discrete-time filter that results from the design process. Hence we normally let $T = 1$ for simplicity. This produces the transfer function

$$H(z) = \frac{0.4716}{1 - e^{-0.4716}z^{-1}} + \frac{-0.236 - 0.136j}{1 - e^{(-0.236+0.408j)}z^{-1}} + \frac{-0.236 + 0.136j}{1 - e^{(-0.236-0.408j)}z^{-1}}$$

Combining we get,

$$H(z) = \frac{0.4716}{1 - 0.624z^{-1}} + \frac{-0.472 + 0.341z^{-1}}{1 - 1.45z^{-1} + 0.624z^{-2}}$$

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