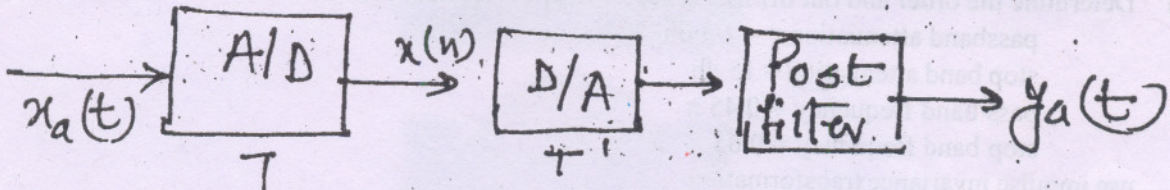


- N.B. :** (1) Question No. 1 is compulsory.
 (2) Attempt any four questions out of remaining six.
 (3) Figures to the right indicates full marks.
 (4) Assume suitable data if necessary.

1. (a) Compare Recursive and non-recursive filter. 20
 (b) Test Linearity and time invariance of following system :—
 (i) $y(n) = a \cos [x(n)] + b$. (ii) $y(n) = (n + 1) x(n)$.
 (c) Using DIF-FFT algorithms find 1 DFT of $x(k) = \{ 10 - 2 + 2j, -2, -2 - 2j \}$.
 (d) Find Z-transform of $x(n) = (n + 1) a^n u(n)$. Specify its ROC.

2. (a) Consider a simple signal processing system as shown in figure. The sampling period of A/D and D/A convertor are $T = 5$ ms and $T = 1$ ms respectively. Determine the output $y_a(t)$ of the if input is $x_a(t) = 3 \cos 100 \pi t + 2 \sin 250 \pi t$. The post filter removes any frequency component above $F_s/2$. 10



- (b) Determine the Convolution of following pairs of signals using Z-transform— 10

(i) $x_1(n) = \left(\frac{1}{4}\right)^n n(n-1)$

$x_2(n) = \left[1 + \left(\frac{1}{2}\right)^n\right] u(n)$

(ii) $x_1(n) = nu(n)$
 $x_2(n) = 2^n u(n-2)$.

3. (a) Show Pole zero diagram with arbitrary pole-zero values for an IIR filter, which has damped Sinusoidal impulse response. Justify your answer. 6
 (b) $x_1(n)$ and $x_2(n)$ are two 8 point real sequences. $x_2(n)$ is time-reversed version of $x_1(n)$. 8

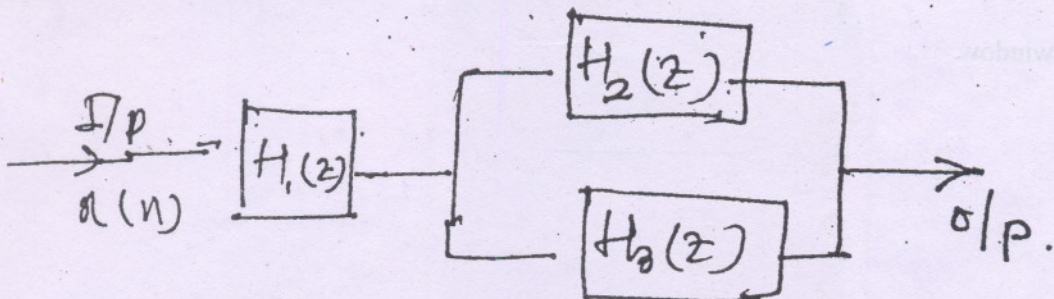
Let $x_1(k) = x_R(k) + x_I(k)$

where R and I represents real and imaging part of DFT.

If $x(n) = x_1(n) + x_2(n)$. Withouts performing any DFT operation, find $x(k)$.

- (c) Let $x_1(n) = [x_0, x_1, x_2, x_3]$ and $x_1(k) = [x_1(0), x_1(1), x_1(2), x_1(3)]$. If $x_2(n) = [x_0, 0, x_1, 0, x_2, 0, x_3, 0]$. 6
 Find $x_2(k)$ using results of $x_1(k)$. State the properly used. Verify the properly.

4. A Discrete time LTI causal system is shown in figure.



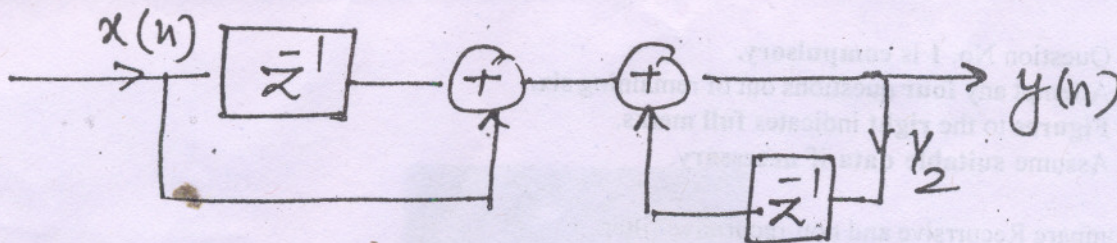
The poles and zeros of individuals modules are tabulated below—

Module	Zero Location	Role Location	Gain
$H_1(z)$	-0.2	-0.4	1
$H_2(z)$	—	-0.2	$-\frac{1}{3}$
$H_3(z)$	—	0.4	$\frac{4}{3}$

- (i) Find transfer function $H(z)$ of total system. 5
 (ii) Find the difference Equation of system. 3
 (iii) Show direct form I, II and parallel form of realisation. 6
 (iv) Find impulse response of system. 3
 (v) Find step response of system. 3

5. (a) Consider the system as shown in figure—

8



- (i) Determine its impulse response $h(n)$.
- (ii) Show that $h(n)$ is the convolution of following signals.

$$h_1(n) = \delta(n) + \delta(n - 1)$$

$$h_2(n) = \left(\frac{1}{2}\right)^2 u(n).$$

- (b) Derive the expression for the order of Butterworth filter.
- (c) Determine the order and cut off frequency of lowpass Butterworth filter if.

4

8

passband attenuation = -1.5 db

stop band attenuation = 15 db

pass band frequency = 0.45π

stop band frequency = 0.65π

use impulse invariance transformation.

6. (a) Consider the following analog sinusoidal signal $x_a(t) = 3 \sin(100 \pi t)$

12

- (i) Sketch the signal for $0 \leq t \leq 30$ ms.
- (ii) The signal is sampled with a sampling period $F_s = 300$ samples/s. Determine the frequency of resulting discrete the signal.
- (iii) Compute the sample value in one period of $x(n)$. Sketch $x(n)$ on the same diagram with $x_a(t)$. What is period of discrete time signal in milisecond.
- (iv) Can you find a sampling rate F_s such that signal reduces to its peak value of 3? What is minimum value of F_s suitable for the same.

(b) Determine zero state and zero-input response for a system.

8

$$y(n) = -0.1 y(n - 1) + 0.2 y(n - 2) + x(n)$$

where $x(n) = (1/3)^n u(n)$ and $y(-1) = y(-2) = 1$.

7. (a) Is the following filter is a Linear phase filter, if—

5

$$H(z) = 1 - z^{-1} + z^{-3} - z^{-4}$$

If yes, draw the phase response to prove it.

(b) Derive a relation between auto correlation of input, impulse response of system and an auto correlation of output.

5

(c) Design a Linear phase FIR filter with the following specifications—

10

$$H_d(w) = 0 \quad 0 \leq |w| \leq \pi/4$$

$$= 2e^{-j3/2w} \quad \frac{\pi}{4} < |w| < \pi.$$

Use Hamming window.