Paper / Subject Code: 32222 / Discrete Time Signal Processing

Duration: 3hrs [Max Marks: 80]

- N.B.: (1) Question No 1 is Compulsory.
 - (2) Attempt any three questions out of the remaining five.
 - (3) All questions carry equal marks.
 - (4) Assume suitable data, if required and state it clearly.
- 1 Attempt any **FOUR**

[20]

a Sketch the frequency response and identify the following filters based on their pass band:

i).
$$h(n) = \left\{1, -\frac{1}{2}\right\}$$
 ii). $H(z) = \frac{z^{-1} - a}{1 - az^{-1}}$

- b Find the IDIF-FFT for a given sequence $X(k) = \{26, -2+2j, -2, -2-2j\}$.
- c Explain the effects of coefficient quantization in FIR filters.
- d Explain frequency warping in bilinear transformation
- e Compare IIM and BLT.
- 2 a Design FIR filter using frequency sampling technique for the following [10] specifications:

$$H(e^{j\omega}) = \begin{cases} e^{-j3\omega} & 0 \le |\omega| \le \frac{\pi}{2} \\ 0 & \text{otherwise} \end{cases}$$

b Design a Butterworth low pass filter using BLT for the following specifications [10]
(Assume T=1sec)

$$0.9 \le |H(e^{j\omega})| \le 1$$
 for $0 \le \omega \le \frac{\pi}{2}$

$$|H(e^{j\omega})| \le 0.1$$
 for $\frac{3\pi}{4} \le \omega \le \pi$

- 3 a If $x(n) = \{1, 2, 3, 4, 5, 6, 7, 8\}$, Find X(k) using DIT-FFT algorithm. Compare [10] the computational complexity of above algorithm with DFT.
 - b The transfer function of digital causal system is given as follows: [10]

$$H(z) = \frac{1 - z^{-1}}{1 - 0.2z^{-1} - 0.15z^{-2}}$$

- (i) Find the difference equation.
- (ii) Draw cascade form, parallel form realization.

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4 a
$$x(n) = \begin{cases} 1, & 0 \le n \le 3 \\ 0, & 4 \le n \le 7 \end{cases}$$
 [10]

- i). Determine the DIF-FFT of the sequence x(n)
- ii). Also find the DFT of the following sequences using the result obtained in (i)

$$x_{1}(n) = \begin{cases} 1, & n = 0 \\ 0, & 1 \le n \le 4 \\ 1, & 5 \le n \le 7 \end{cases}$$

$$x_{2}(n) = \begin{cases} 0, & 0 \le n \le 1 \\ 1, & 2 \le n \le 5 \\ 0, & 6 < n < 7 \end{cases}$$

b Design a FIR filter using window method for following specification. [10]

$$H(e^{j\omega}) = \begin{cases} e^{-j2\omega} & for -\frac{\pi}{4} \le \omega \le \frac{\pi}{4} \\ 0 & for \frac{\pi}{4} \le |\omega| \le \pi \end{cases}$$

Determine the filter coefficient h(n) if the window function is defined as

$$w[n] = \begin{cases} 1 & 0 \le n \le 4 \\ 0 & otherwise \end{cases}$$

Also determine the frequency response of the designed filter.

5 a An analog filter has transfer function

$$H(s) = \frac{s + 0.1}{(s + 0.1)^2 + 16}$$

[10]

Determine the transfer function of digital filter using bilinear transformation.

The digital filter should have specification $\omega_r = \frac{\pi}{2}$

b The unit sample response of a system is $h(n) = \{1, 2\}$ use overlap-save method of linear filtering to determine output sequence for the repeating input sequence,

$$x(n) = \{1, 2, 3, 4, 5, 6, 7, 8, 9, 10\}$$
. Take N=5

- 6 a Explain application of DSP in ECG and EEG signal analysis. [10]
 - b Obtain lattice realization for FIR filter given by [10]

$$H(z) = 1 + \frac{3}{4}z^{-1} + \frac{1}{2}z^{-2} + \frac{1}{4}z^{-3}$$
