

CBCS SCHEME

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18EI/BM52

Fifth Semester B.E. Degree Examination, Jan./Feb. 2021 Fundamentals of Signals and DSP

Time: 3 hrs.

Max. Marks: 100

Note: Answer any FIVE full questions, choosing ONE full question from each module.

Module-1

- 1 a. With necessary diagram, briefly describe the difference between Analog and Digital Signal Processing system. Mention any two advantages of digital system over analog system. (08 Marks)
- b. State and explain the Sampling theorem, with necessary waveforms. (08 Marks)
- c. Consider the analog signal $x_a(t) = 3\cos 50\pi t + 10\sin 3000\pi t - \cos 100\pi t$. Find the Nyquist rate for the signal. (04 Marks)

OR

- 2 a. Consider the Discrete time signal $x_1(n)$ and $x_2(n)$ as shown in Fig.Q2(a(i)) & Q2(a(ii)). Sketch and label the following signal.
 - i) $y_1(n) = 2x_2(n)$
 - ii) $y_2(n) = x_1(n) - x_2(n)$
 - iii) $y_3(n) = x_1(n) x_2(n)$. (08 Marks)

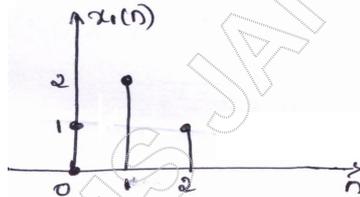


Fig. Q2(a(i))

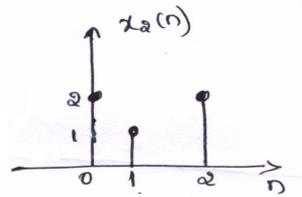


Fig. Q2(a(ii))

- b. Determine whether the following systems are causal (or) non-causal.
 - i) $y(n) = x(n) - x(n-1)$
 - ii) $y(n) = 3y(n-1) - nx(n) + 5x(n-1) - 6x(n-1)$
 - iii) $y(n) = y(n)x(n)$
 - iv) $y(n) = e^{x(n)}$. (08 Marks)
- c. Find the convolution of sequence using Tabular method
 $x_1(n) = \{4, 1, 1\}$ $x_2(n) = \{2, 1, 1\}$. (04 Marks)

Module-2

- 3 a. Find the Z-transform of the following : i) $x(n) = u(-n)$ ii) $x(n) = -\beta^n u(-n-1)$
 iii) $x(n) = 6\alpha^n u(n)$ iv) $x(n) = n u(n)$. (08 Marks)
- b. Use Z-transform to perform the convolution of the following two sequences.
 $h(n) = \delta(n) + 4\delta(n-1)$ and $x(n) = \delta(n) - 5\delta(n-1)$. (08 Marks)
- c. Prove $Z[x(n+1)] = Zx(z) - Zx(0)$. (04 Marks)

OR

- 4 a. Find the Region of convergence and Z transform for the sequence.

$$x(n) = \begin{cases} 4^n & \text{for } n \geq 0 \\ 5^{-n} & \text{for } n < 0 \end{cases}$$
 (08 Marks)
- b. Find the inverse Z-transform of the sequence

$$X(z) = \frac{Z}{3z^2 - 4z + 1}$$
 for the following ROC's i) $|z| > 1$ ii) $\frac{1}{3} < |z| < 1$. (08 Marks)

Important Note : 1. On completing your answers, compulsorily draw diagonal cross lines on the remaining blank pages.
2. Any revealing of identification, appeal to evaluator and/or equations written eg, 42+8=50, will be treated as malpractice.

- c. Consider a difference equation $y(n) - \frac{3}{4}y(n-1) + \frac{1}{8}y(n-2) = x(n) + \frac{5}{3}x(n-1)$. Realize the system using direct form - I. (04 Marks)

Module-3

- 5 a. Define DFT and find 8 point DFT of the sequence $x(n) = \{1, 1, 1, 1, 1, 1, 0, 0\}$. (10 Marks)
 b. Explain the following properties of DFT : i) Periodicity ii) Linearity iii) Parseval's theorem. (10 Marks)

OR

- 6 a. Find the N point DFT of the sequence :
 i) $x(n) = e^{j\omega n}$, $0 \leq n \leq N-1$ ii) $x(n) = 1$, $0 \leq n \leq N-1$. (10 Marks)
 b. Compute $X(k)$ for sequence $x(n) = 2^n$ using 8 point DIT FFT algorithm. (10 Marks)

Module-4

- 7 a. List any three advantages and disadvantages of IIR filters. (06 Marks)
 b. Design Low pass Butterworth filter using Bilinear transformation method for satisfying the following constraints : i) Pass band $\omega_p : 0.162$ rad ii) Stop band $\omega_s : 1.63$ rad iii) Pass band ripple : 3 dB iv) Stop band attenuation : 30dB v) Sampling frequency : 8 KHz. (12 Marks)
 c. Briefly describe difference between the Butterworth and Chebyshev filter. (02 Marks)

OR

- 8 a. Write the magnitude response of Hamming, Hanning and Rectangular windows. (06 Marks)
 b. Design low pass filter for the following desired frequency response using Hamming and Hanning window.

$$H_d(e^{j\omega}) = \begin{cases} e^{-j\omega 3} & 0 \leq |\omega| \leq \frac{\pi}{3} \\ 0 & \frac{\pi}{3} \leq |\omega| \leq \pi \end{cases} \quad (12 \text{ Marks})$$

- c. Mention any two advantages of FIR filter. (02 Marks)

Module-5

- 9 a. With required diagram, explain the decimation process. (08 Marks)
 b. With suitable diagram, explain Digital and Quadrature mirror filter bank. (08 Marks)
 c. Consider a sample sequence $x(n) = \{0, 3, 6, 9, 12\}$, using linear interpolation method increases the sampling rate for $L = 2$. (04 Marks)

OR

- 10 a. With neat diagram and necessary equation, explain Adaptive filters. (08 Marks)
 b. Explain the following applications of Adaptive filtering :
 i) Noise Reduction for the hearing impaired ii) Fetal ECG from Abdominal ECG. (08 Marks)
 c. Obtain the decimated signal $w(n)$ by a factor 3 from the input signal.
 $x(n) = \{0, 5, 4, 3, 1, 2, 5, 0, 5, 4, 3, 1, 2, 5, 0\}$. Obtain output $y(n)$ by interpolating the signal by a factor 3 and what is linear interpolation. (04 Marks)