

# 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)

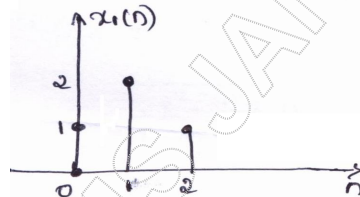


Fig. Q2(a(i))

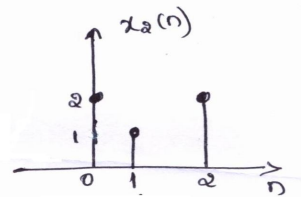


Fig. Q2(a(ii))

- 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)

- 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)