

**GUJARAT TECHNOLOGICAL UNIVERSITY****BE - SEMESTER-VII (NEW) EXAMINATION – WINTER 2022****Subject Code:2171003****Date:16-01-2023****Subject Name:Digital Signal Processing****Time:10:30 AM TO 01:00 PM****Total Marks:70****Instructions:**

1. Attempt all questions.
2. Make suitable assumptions wherever necessary.
3. Figures to the right indicate full marks.
4. Simple and non-programmable scientific calculators are allowed.

		MARKS
<b>Q.1</b>	(a) Represent the sequence $x(n) = \{0.5, 3, 0, 1\}$ into a sum of weighted impulse sequences.	<b>03</b>
	(b) What is Nyquist criteria for sampling? What will happen if sampling process do not follow the Nyquist criteria?	<b>04</b>
	(c) The impulse response of a linear time-invariant system is $h(n) = \{1, 2, 1, 0, 1\}$ . Determine by graphical method, response of the <div style="margin-left: 40px;"> <math>\uparrow</math>            System to the input signal <math>x(n) = \{1, 2, 3, 1\}</math>.         </div>	<b>07</b>
<b>Q.2</b>	(a) A discrete-time signal $x(n)$ is defined as:	<b>03</b>
	$x(n) = \begin{cases} 1 + \frac{n}{3}, & -3 \leq n \leq -1 \\ 1, & 0 \leq n \leq 3 \\ 0, & \text{elsewhere} \end{cases}$	
	Determine its values and sketch the signal $x(n)$ .	
	(b) State and prove frequency shifting and convolution properties of z-transform.	<b>04</b>
	(c) Consider a system with impulse response	<b>07</b>
	$h(n) = \begin{cases} \left(\frac{1}{2}\right)^n, & 0 \leq n \leq 4 \\ 0, & \text{elsewhere} \end{cases}$	
	Determine the input $x(n)$ for $0 \leq n \leq 8$ that will generate the output sequence $y(n) = \{1, 2, 2.5, 3, 3, 3, 2, 1, 0, \dots\}$	
	<b>OR</b>	
	(c) A LTI system is characterized by the system function	<b>07</b>
	$H(Z) = \frac{1}{1 - \frac{1}{2}z^{-1}} + \frac{2}{1 - 3z^{-1}}$	
	Specify the ROC of $H(z)$ and determine $h(n)$ for the following conditions:	
	i) The system is stable.	
	ii) The system is causal.	
	iii) The system is anticausal.	
<b>Q.3</b>	(a) Draw nature of Region of Convergence (ROC) in Z-plane for infinite duration causal, anticausal and two-sided sequence.	<b>03</b>
	(b) Determine the response of the system	<b>04</b>

$$y(n) = \frac{5}{6} y(n-1) - \frac{1}{6} y(n-2) + x(n)$$

to the input signal  $x(n) = \delta(n) - \frac{1}{3}\delta(n-1)$ .

- (c) Obtain the direct form I and direct form II structures for the following system: **07**

$$H(Z) = \frac{1 + 0.875 z^{-1}}{(1 + 0.2 z^{-1} + 0.9 z^{-2})(1 - 0.7 z^{-1})}$$

**OR**

- Q.3** (a) Consider the signal **03**

$$x(n) = \{-1, 2, -3, 2, -1\}$$

↑

with Fourier transform  $X(\omega)$ . Compute  $\int_{-\pi}^{\pi} |X(\omega)|^2 d\omega$ .

- (b) Compute the 4-point DFT of a discrete time sequence  $\{1, 0, 2, 1\}$ . **04**  
 (c) Determine the zeros for the following FIR systems and indicate whether the system is minimum phase, maximum phase, or mixed phase. **07**

$$H_1(Z) = 6 + z^{-1} - z^{-2}$$

$$H_2(Z) = 1 - z^{-1} - 6z^{-2}$$

$$H_3(Z) = 1 - \frac{5}{2}z^{-1} - \frac{3}{2}z^{-2}$$

$$H_4(Z) = 1 + \frac{5}{3}z^{-1} - \frac{2}{3}z^{-2}$$

- Q.4** (a) Explain how circular convolution can be converted to linear convolution by zero padding. **03**  
 (b) Using decimation in time algorithm, compute 4-point DFT of the sequence  $x(n) = \{0, 1, 2, 3\}$  **04**  
 (c) 4-point DFT of a sequence  $x(n)$  is  $\{1, 0, 1, 0\}$ . By using butterfly diagram as in DIT-FFT, find  $x(n)$ . **07**

**OR**

- Q.4** (a) Explain necessity of windowing in FIR filter design. **03**  
 (b) Find out  $H(Z)$  for the given  $H(S) = \frac{2}{s^2 + 3s + 2}$  using impulse invariance method. Assume  $T=1s$ . **04**  
 (c) Consider input sequence  $x(n) = \{1, 2, 3\}$  and impulse response of a system  $h(n) = \{1, 1\}$ . Find the linear convolution using graphical circular convolution method. Match result of same using tabulation/ matrix method. **07**

- Q.5** (a) Explain need of anti-aliasing filter in a down sampler. **03**  
 (b) Explain how to achieve sampling rate conversion by rational factor  $L/M$ . Where  $L$  and  $M$  are constant. **04**  
 (c) List and explain the application of Multirate Signal Processing. **07**

**OR**

- Q.5** (a) Explain how echo cancellation is achieved using adaptive filters. **03**  
 (b) Compare Up sampling and Down Sampling. **04**  
 (c) Explain pipelining and MAC architecture of Digital Signal Processor. **07**

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