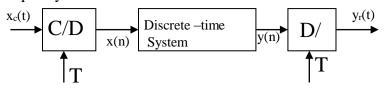
GUJARAT TECHNOLOGICAL UNIVERSITY BE - SEMESTER-VII • EXAMINATION – WINTER • 2014

$\mathbf{DE} = \mathbf{SER} = \mathbf{V} \mathbf{H} = \mathbf{E} \mathbf{A} \mathbf{V} \mathbf{H} \mathbf{A} \mathbf{H} \mathbf{O} \mathbf{N} = \mathbf{V} \mathbf{H} \mathbf{V} \mathbf{E} \mathbf{K} = \mathbf{V} \mathbf{H} \mathbf{V} \mathbf{E} \mathbf{K} + \mathbf{V} \mathbf{H} \mathbf{H} \mathbf{H} \mathbf{H} \mathbf{H} \mathbf{H} \mathbf{H} H$			
•		e: 171003 Date: 04-12-2014	
Subject Name: Digital Signal Processing			
Time: 10:30 am - 01:00 pm Total Marks: 7			
Instructions:			
1. Attempt all questions.			
	 Make suitable assumptions wherever necessary. Figure to the night in directs full models. 		
3.	rigu	res to the right indicate full marks.	
Q.1	(a)	Draw the block diagram of a typical Digital Signal Processing system and explain.	07
	(b)		07
		$\mathbf{x}(\mathbf{n}) = \{ 1, 1, 1, 1, 1, \frac{1}{2} \}$	
		Sketch and label carefully each of the following signals: (i)r(r, 2) $(iii) r(4, r)$ $(iii) r(2r)$	
		(i) $x(n-2)$ (ii) $x(4-n)$ (iii) $x(2n)$ (iv) $x(n)u(2-n)$ (v) $x(n-1)\delta(n-3)$	
Q.2	(n)	Perform the linear convolution of the following sequences :	07
Q.2	(a)	$x_1(n) = \{1, 2, 3, 4, 5\}, x_2(n) = \{-1, 0, 1\}$	07
		$\uparrow \qquad \uparrow \qquad \uparrow$	
	(b)	(i) For the following system, determine whether the system is	05
	. ,	stable, causal, linear, time-invariant, memoryless:	
		n	
		$T{x(n)} = \Sigma x(k)$	
		$k=n_0$ (ii) What are the advantages of digital signal processing over analog signal	02
		processing?	02
		OR	
	(b)	Let $X(e^{iw})$ denote the fourier transform of the signal $x(n)$. Perform the following	07
	(~)	calculations without explicitly evaluating $X(e^{jw})$:	0.
		$x(n) = \{-1, 0, 1, 2, 1, 0, 1, 2, 1, 0, -1\}$	
		(i) Evaluate $X(e^{iw}) \mid w=0$	
		(ii) Evaluate $X(e^{jw}) \mid w=\pi$	
		(iii) Find $\Theta(X(e^{jw}))$	
		π	
		(iv) Evaluate $\int X(e^{jw}) dw$	
		-π	
		(v) Determine and sketch the signal whose fourier transform is $X(e^{J^W})$	
Q.3	(a)	(vi) Determine and sketch the signal whose fourier transform is $Re{X(e^{jw})}$ Determine the z-transform of the following sequences. Sketch ROC and pole	07
Q.5	(a)	zero plot :	07
		$\begin{array}{c} \text{(i)} x_1(n) = \alpha^{ n }, 0 < \alpha < 1 \end{array}$	
		$(ii)x_2(n) = (-1/3)^n u(n) - (1/2)^n u(-n-1)$	
	(b)	Suppose the z-transform of $x(n)$ is	07
		$X(z) = \frac{z^{10}}{(z-(1/2))(z-(3/2))^{10}(z+(3/2))^2(z+(5/2))(z+(7/2))}$	
		$(z-(1/2)) (z-(3/2))^{10} (z + (3/2))^2 (z + (5/2)) (z + (7/2))$	
		It is also known that $x(n)$ is a stable sequence.	
		(i)Determine the region of convergence of $X(z)$. (ii) Determine $y(z)$ of $z = -8$	
		(ii) Determine $x(n)$ at $n = -8$. OR	
Q.3	(a)	Consider the discrete time system with an ideal low pass filter with cutoff	07
X	()		1 ()

frequency $\pi/8$ radian/s.



(i)If $x_{c}(t)$ is bandlimited to 5 kHz , what is the maximum value of T that will avoid aliasing?

(ii)If 1/T = 10 kHz, what will the cutoff frequency of the continuous –time filter be?

(iii) Repeat part(ii) for 1/T=20kHz. (b) Draw the structures of the following discrete time system: 07 $H(z) = (1 + z^{-1})^2$ 1- $0.75 z^{-1} + 0.125 z^{-2}$ (i)Direct form – I (ii)Direct Form – II (iii)Cascade form (iv)Parallel form (a) Discuss the following transformation methods to design digital filters: 07 **Q.4** (i)Impulse invariance (ii)Bilinear transformation **(b)** Find the circular convolution of the following sequences: 07 $x_1(n) = \{1, 2, 3, 4\}$ $x_2(n) = \{2, 1, 2, 1\}$ 1 1 OR Design a Digital low pass FIR filter using Kaiser window to meet the 07 **Q.4** (a) following specifications: $0.99 \le |H(e^{jw})| \le 1.01$, $0 \le w \le 0.4\pi$ $|H(e^{jw})| < 0.001$, $0.6 \pi < w < \pi$ Consider the real finite-length sequence x(n). 07 **(b)** $x(n) = \{4, 3, 2, 1\}$ 1 (i)Sketch the finite length sequence y(n) whose six-point DFT is Y(k) = $W_6^{4k} X(k)$, Where X(k) is the six-point DFT of x(n).

(ii) Sketch the finite length sequence w(n) whose six-point DFT is $W(k) = Re\{X(k)\}$

(iii) Sketch the finite length sequence q(n) whose three-point DFT is Q(k) = X(2k), k=0,1,2

- Q.5 (a) Explain the Decimation in Time FFT algorithm 07
 (b) Discuss the applications of digital signal processing with suitable examples. 07
 OR
 Q.5 (a) Discuss the key features of the architecture of DSP Processors. 07
 (b) Write a short note on coefficient quantization in IIR filters. 07
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