Reg.	No.

Question Paper Code : 57318

B.E./B.Tech. DEGREE EXAMINATION, MAY/JUNE 2016

Fourth Semester

Electrical and Electronics Engineering

EE 6403 – DISCRETE TIME SYSTEMS AND SIGNAL PROCESSING

(Common to Instrumentation and Control Engineering, Electronics and Instrumentation Engineering)

(Regulations 2013)

Time : Three Hours

Maximum : 100 Marks

Answer ALL questions. PART – A $(10 \times 2 = 20 \text{ Marks})$

- 1. Determine if the system described by the equation $y(n) = x(n) + \frac{1}{x(n-1)}$ is causal or non-causal.
- 2. What is an Anti-Aliasing filter?
- Determine the Z-transform and ROC of the following finite duration signals
 (i) x(n) = {3, 2, 2, 3, 5, 0, 1}
 - (ii) $x(n) = \delta(n-k)$
- Compute the convolution of the two sequences
 x(n) = {2, 1, 0, 0.5} and n(n) = {2, 2, 1, 1}
- 5. Draw the flow graph of a 4 point radix-2 DIT-FFT butterfly structure for DFT.
- 6. What are the applications of FFT algorithm ?

7. Obtain the cascade realization for the system function,

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

- 8. Mention the advantages of FIR filters over IIR filters.
- 9. What are the merits and demerits of VLIW architecture?
- 10. What are the factors that influence the selection of DSP processor for an application?

$PART - B (5 \times 16 = 80 Marks)$

11. (a) (i) Determine if the signals, $x_1(n)$ and $x_2(n)$ are power, energy or neither energy nor power signals.

$$x_1(n) = \left(\frac{1}{3}\right)^n u(n) \text{ and } x_2(n) = e^{2n} u(n).$$
 (8)

(ii) What is the input signal x(n) that will generate the output sequence

 $y(n) = \{1, 5, 10, 11, 8, 4, 1\}$ for a system with impulse response $h(n) = \{1, 2, 1\}.$ (8)

(6)

(5)

(5)

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OR

- (b) (i) A signal x(t) = sin c(50 πt) is sampled at a rate of (1) 20 Hz (2) 50 Hz and
 (3) 75 Hz. For each of these cases, explain if you can recover the signal x(t) from the samples signal.
 - (ii) Determine whether or not each of the following signals is periodic. If the signal is periodic, specify its fundamental period.

(1)
$$x(n) = e^{j6\pi n}$$

(2)
$$x(n) = \cos \frac{\pi}{3}n + \cos \frac{3\pi}{4}n$$

2

12. (a) (i)

F

ind x(n) if X(z) =
$$\frac{1 + \frac{1}{2}z^{-1}}{1 - \frac{1}{2}z^{-1}}$$

(ii) Find the response of the causal system y(n) - y(n-1) = x(n) + x(n-1) to the input x(n) = u(n). Test its stability. (10)

OR

(b) Find the impulse response, frequency response, magnitude response and phase response of the second order system.

$$y(n) - y(n-1) + \frac{3}{16}y(n-2) = x(n) - \frac{1}{2}x(n-1).$$
 (16)

- 13. (a) (i) Summarize the steps of radix -2 DIT-FFT algorithm. (8)
 - (ii) Compute the 4 point DFT of the sequence x(n) = {0, 1, 2, 3} using DIT and DIF algorithm.
 (8)

OR

 $X(K) = \{4, 1 - j 2.414, 0, 1 - j 0.414, 0, 1 + j 0.414, 0, 1 + j 2.414\}$ Using DIF algorithm.

14. (a) Design an ideal low pass filter with a frequency response

$$H_d(e^{jw}) = 1 \text{ for } \frac{-\pi}{2} \le w \le \frac{\pi}{2}$$

$$= 0 \text{ for } \frac{\pi}{2} \le |\mathbf{w}| \le \pi$$

Find the values of h(n) for N = 11. Find H(z) and the filter coefficients. (16)

OR

(b) (i) Given the specifications $\alpha_p = 3 \text{ dB}$, $\alpha_s = 10 \text{ dB}$, $f_p = 1 \text{ kHz}$ and $f_s = 2 \text{ kHz}$. Determine the order of the filter using Chebyshev approximation. Find H(s). (8)

(ii) Apply bilinear transformation to

$$H(s) = \frac{2}{(s+1)(s+2)}$$
 with T = 1 sec and find H(z). (8)

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(16)

(6)

- 15. (a) (i) Discuss on the addressing modes supported by a DSP processor.
 - (ii) Design a DSP based system for the process of Audio signals in an audio recorder system.
 (8)

OR

(b) (i) Explain the datapath architecture and the bus structure in a DSP processor with suitable diagram.
 (8)

Summarize the steps of radiv 12 DIT-IT 4 sponthing

(ii) Elaborate on Radar signal processing using a DSP processor.

(8)

(8)