

Reg. No.		/	 	 	 		 	,	
	Reg. No.					:			

Question Paper Code: 71039

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

Fifth Semester

Electrical and Electronics Engineering

EC 65 – DIGITAL SIGNAL PROCESSING

(Common to Sixth Semester Electronics and Instrumentation Engineering and Instrumentation and Control Engineering)

(Regulations 2008)

Time: Three Hours

Maximum: 100 Marks

Answer ALL questions. $PART - A (10 \times 2 = 20 Marks)$

- 1. Given a continuous time signal $x(t) = 2 \cos 500 \pi l$. What is the Nyquist rate and fundamental frequency of the signal?
- 2. Determine whether x[n] = u[n] is a power signal or an energy signal.
- 3. What is the Z transform of $x(n) = \delta(n) 0.95 \delta(n-1)$?
- 4. State the convolution property of Z transform.
- 5. Draw the butterfly diagram for decimation in time FFT algorithm.
- 6. What is the relationship between DFT and Z transform?
- 7. Define pre-warping effect. Why is it employed?
- 8. Give Hamming window function.
- 9. What is meant by bit reversed addressing mode? What is the application for which this addressing mode is preferred?
- 10. Compare the RISC and CISC processors.

15-06

1

71039

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

11. (a) Compute the convolution for the signals

(16)

$$x(n) = \alpha^n, -3 \le n \le 5$$

0, elsewhere

$$h(n) = 1, \quad 0 \le n \le 4$$

0, elsewhere

OR

(b) Consider the analog signal

 $x_a(t) = 3 \cos 2000 \pi t + 5 \sin 6000 \pi t = 10 \cos 12,000 \pi t$

(i) What is the Nyquist rate for this signal?

- (4)
- (ii) Assume now that we sample this signal using a sampling rate $F_S = 5000$ samples/sec. What is the discrete time signal obtained after sampling? (6)
- (iii) What is the analog signal y_a(t) that we can reconstruct from the samples if we use ideal interpolation. (6)
- 12. (a) (i) Find the Z-transform and its associated ROC for the following discrete time signal $x[n] = \left(\frac{-1}{5}\right)^n u[n] + 5\left(\frac{1}{2}\right)^{-n} u[n-1]$. (8)
 - (ii) Evaluate the frequency response of the system described by system function $H(z) = \frac{1}{1 0.5z^{-1}}$. (8)

OR

(b) Using z-transform determine the response y[n] for $n \ge 0$ if

$$y[n] = \frac{1}{2}y[n-1] + x[n], x[n] = \left(\frac{1}{3}\right)^n u(n) y(-1) = 1.$$
 (16)

- 13. (a) (i) The first five points of the eight point DFT of a real valued sequence are $\{0.25, 0.125 j0.3018, 0, 0.125 j0.0518, 0\}$. Determine the remaining three points. (4)
 - (ii) Compute the eight point DFT of the sequence x = [1, 1, 1, 1, 1, 1, 1, 1], using Decimation-in-frequency FFT algorithm. (12)

OR

(b) Consider the sequences:

$$x_1(n) = \{0, 1, 2, 3, 4\}, x_2(n) = \{0, 1, 0, 0, 0\}$$

$$s(n) = \{1, 0, 0, 0, 0\}$$

(i) Determine a sequence
$$y(n)$$
 so that $Y(k) = X_1(k) X_2(k)$. (8)

(ii) Is there a sequence
$$x_3(n)$$
 such that $S(k) = X_1(k) X_3(k)$? (8)

14. (a) Design an FIR linear phase, digital filter approximating the ideal frequency

response
$$H_d(\omega) = \begin{cases} 1, & |\omega| \le \frac{\pi}{6} \\ 0, & \frac{\pi}{6} < |\omega| \le \pi \end{cases}$$

Determine the coefficients of a 25 tap filter based on the window method with a rectangular window. (16)

OR

- (b) (i) Convert the analog filter with system function $H_a(s) = \frac{s + 0.1}{(s + 0.1)^2 + 9}$ into a digital IIR filter by means of the impulse invariance method. (8)
 - (ii) Draw the direct form I and direct form II structures for the given difference equation y(n) = y(n-1) 0.5 y(n-2) + x(n) x(n-1) + x(n+2). (8)
- 15. (a) Explain Von Neumann, Harvard architecture and modified Harvard architecture for the computer. (16)

OR

- (b) (i) Explain how convolution is performed using a single MAC unit. (8)
 - (ii) Discuss the addressing modes used in programmable DSPs. (8)