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**Question Paper Code : 11039**

B.E./B.Tech. DEGREE EXAMINATION, NOVEMBER/DECEMBER 2015

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 × 2 = 20 marks)

1. Given a continuous time signal  $x(t) = 2 \cos 500\pi t$ . 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. Determine the z-transform and ROC for the signal  $x(n) = \delta(n - k) + \delta(n + k)$ .
4. Prove the convolution property of z-transform.
5. Draw the basic butterfly of DIF-FFT.
6. What is the relationship between DFT and Z transform?
7. Define pre-warping effect. Why it is employed?
8. Give Hamming window function.
9. Define periodogram.
10. Define Gibbs phenomena.

PART B — (5 × 16 = 80 marks)

11. (a) Compute the convolution for the signals (16)

$$x(n) = \begin{cases} a^n, & -3 \leq n \leq 5 \\ 0, & \text{elsewhere} \end{cases}$$

$$h(n) = \begin{cases} 1, & 0 \leq n \leq 4 \\ 0, & \text{elsewhere.} \end{cases}$$

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 ROC of  $x(n) = r^2 \cos(n\theta)u(n)$ . (8)
- (ii) Find Inverse Z — Transform of  $X(z) = z/[3z^2 - 4z + 1]$ , ROC  $|z| > 1$ . (8)

Or

- (b) (i) Determine the DTFT of the given sequence  $x[n] = a^n(u(n) - u(n-8))$ ,  $|a| < 1$ . (8)
- (ii) Prove the linearity and frequency shifting theorem of the Discrete Time Fourier Transform. (8)
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 a low pass Chebychev filter with an acceptable passband ripple of 2 dB, cut off frequency of 1 rad/sec. and stop band attenuation of 20 dB or greater beyond 1.3 rad/sec. (16)

Or

(b) Design a low pass FIR filter using the Hanning window for which the desired frequency response is

$$H_d(\omega) = \begin{cases} e^{-j\omega\alpha}, & |\omega| = \omega_c \\ 0, & \text{elsewhere} \end{cases}$$

The length of filter is 7 and  $\omega_c = 1$  rad/sec. (16)

15. (a) Explain various addressing modes of a digital signal processor. (16)

Or

(b) Draw the functional block diagram of a digital signal processor and explain. (16)