Roll No. 3322 976

24792

B.Tech. 6th Sem. (Electronics & Computer Engg.) Examination – May, 2014

DIGITAL SIGNAL PROCESSING

Paper: EC-614-F

| | Lupus | | |
|----------------|--------------------------------|-------------------------------|-----|
| Time : TI | rree hours] | [Maximum Marks : 1 | 00 |
| Before ans | swering the questions, cand | lidates should ensure that th | ıey |
| have been | supplied the correct and | complete question paper. I | No |
| complain | t in this regard, will be ente | rtained after examination. | |
| Note: | Question number 1 is | compulsory, and attem | pt |
| | one question from each | ch of the four sections. A | All |
| | questions carry equal r | narks. | |
| 1 . (a) | Give the steps in the d | esign of a digital filter fro | m |
| | analog filter? | | 3 |
| (Jay) | State the properties of | FIR filter. | 2 |
| (c) | Are FIR filters inheren | tly stable ? | 2 |
| (d) | Give the desirable char | acteristics of the window? | ? 2 |
| | | | |

24792-100-(P-4)(Q-9)(14)

P. T. O.

| / | |
|--|----------|
| (e) Write any four applications of multi-rate sign processing. | na |
| (f) Explain the shifting and scaling property of | - |
| transform with suitable example. | 4 |
| (g) What do you understand by aliasing? How w | ril 3 |
| SECTION - A | |
| 2. (a) What do you understand by signal ? Expla | ir |
| various types of signals. | 8 |
| (b) State and prove the parseval's energy theorem for | or |
| discrete time signal | 2 |
| 3. (a) What do you mean by random and determinist signals ? Explain you answer with suitab | ic le |
| example. | 8 |
| (b) Find Fourier transform of $f(t) = e^{st}$ cosbt. | 2 |
| SECTION - B | |
| (a) What is an IIR filter? Compare an IIR filter with a FIR filter. | in 8 |
| (b) Obtain the direct form I and direct form | П |
| realisations for third order IIR transfer function | m |
| | 2 |
| $H(z) = (0.28z^2 + 0.319z + 0.04)/(0.5z^3 + 0.3z^2 + 0.17z - 0.2)$ |) |
| 24792-100-(P-4)(Q-9)(14) (2) | |

5. (a) A digital communication link carries binary coded words representing samples of an input signal

 $Xq(t) = 3\cos 600\pi t + 2\cos 1800.t$

the link is operated at 10,000 bits/s and each input samples is quantized 1024 different voltage levels.

- (i) What are the sampling frequency and the folding frequency?
- (ii) What is the Nyquist rate for the signal $\chi_q(t)$? 12
- (b) How will you reconstruct band limited signal from its samples? Explain with mathematical equations.

SECTION - C

6. (a) Using residue method, determine x(n) for

(X(z)=1/(z-0.25)(z-0.5), ROC: |Z| > 0.5 12

- (b) Explain final value and initial value theorem. 8
- 7. (a) Using a rectangular window technique design a low pass filter with pass band gain unity, cut off frequency of 1000 Hz & working at a sampling frequency of 5 Khz. The length of the impulse response should be 7.

24792-100-(P-4)(Q-9)(14)

(3)

P. T. O.

(b) Explain the finite word length effect in FIR digital filter.

SECTION - D

- 8. (a) Explain the implementation steps in speech coding using transform coding.
 - (b) Discuss the design steps involved in the implementation of multistage sampling rate converter.
- 9. Explain the efficient implementation of polyphase decimator and interpolator20