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Question Paper Code :

**ANIL NEERUKONDA INSTITUTE OF TECHNOLOGY & SCIENCES
(AUTONOMOUS)**

M.E/M.Tech I-Semester Regular Examinations, November 2015

ADVANCED DIGITAL SIGNAL PROCESSING

(Communication Systems)

Date:

Time: 3 hours

Max Marks: 60

Answer ONE Question from each Unit

All Questions Carry Equal Marks

All parts of the question must be answered in one place only

UNIT-I

1. a. Compare FIR and IIR filters (CO-1,PO-1,3,4) (6M)
b. Design i) butterworth ii) chebyshev analog lowpass filters that have -3dB cutoff frequency of 100 rad/sec and a stop band attenuation of 25dB or greater for radian frequencies past 250 rad/sec. (CO-1,PO-1,3,4) (6M)

(or)

2. a Explain the Remez Algorithm, Parks-McClellan Algorithm (4M)
b. Describe and explain the Optimization methods for designing IIR filters(CO-1,4, PO-1,3,4,8) (8M)

UNIT-II

3. a. Explain about aliasing effect and imaging with neat frequency domain representation (4M)
b. Decimating $x(n)$ by a factor of $D=2$ produces the signal $x_d(n)=x(2n)$ for all n . Show that $x_d(w)=x_s(w/2)$. Plot the signal $x_d(n)$ and its transform $x_d(w)$. Do we decimated the sampled signal $x_s(n)$ (4M)
c. write about the filters used in sampling rate alteration devices(CO-2,4, PO-1,3,4)(4M)

(or)

4. a. Describe the subband coding of speech signals in detail. (4M)
b. Explain about the analysis and synthesis of Digital filter banks(CO-2,4, PO-1,3,4)(8M)

UNIT-III

5. a. Define AR,MA,ARMA processes with respect to rational system function $H(Z)$. (8M)
b. what are the various parameters required to implement forward linear prediction and how they are obtained(CO-2,3,4, PO-1,3,4) (4M)

(or)

6. a. Explain the relationship between forward and backward linear filtering (4M)
b. Explain about the wiener filter for filtering on prediction. (CO-2,3,4, PO-1,3,4) (4M)

UNIT-IV

7. a. Explain the Bluestein algorithm(CO-2,4, PO-1,3,4) (4M)
b. Discuss about the tunable digital filters (5M)
c. Give the relation between Levinson- Durbin algorithm and Schur Algorithm (3M)

(or)

8. a. Explain about Chirp-Z algorithm with neat diagrams. (CO-2,4, PO-1,3,4) (8M)
b. How is the chirp-Z algorithm different from Goertzel algorithm. (4M)

UNIT-V

9. a. Explain about the TDM to FDM converter in detail (CO-4,5, PO-1,3,4,7.12) (4M)
b. Explain briefly about Speech analysis and synthesis. (8M)

(or)

10. a. What is a channel Vocoder.Explain its operation. (5M)
b. Explain about the 10MHz pipelining convolver. (7M)