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# AQ - 2922

# First Semester M. E. (Electronics and Tele.) Full Time (CGS) Examination

# DIGITAL SIGNAL PROCESSING AND APPLICATIONS

## 1 ENTC 4

#### P. Pages : 3

Time : Three Hours ]

[Max. Marks : 80]

- Note : (1) Separate answer book must be used for each section in the subject Geology, Engineering material of civil branch and Separate answer book must be used for Section A and B in Pharmacy and Cosmetic Tech.
  - (2) Answer Three questions from Section A and Three questions from Section B.
  - (3) Due credit will be given to neatness and adequate dimensions.
  - (4) Assume suitable data wherever necessary.
  - (5) Illustrate your answer wherever necessary with the help of neat sketches.
  - (6) All questions carry marks as indicated.
  - (7) Use pen of Blue/Black ink/refill only for writing the answer book.

# SECTION A

1. (a) Explain the following terms as applied to analogue and digital filters :

- (i) Phase delay.
- (ii) Group delay.
- (iii) Linear phase.
- (iv) Minimum phase.
- (b) Explain why discrete time FIR filters can be made exactly linear phase whereas analogue and digital filters with infinite impulse responses cannot.

#### OR

- 2. (a) Distinguish between frequency response of Chebyshev Type-1 and Type-II filter.
  - (b) Derive the expression to determine the poles of Butterworth filter. 7 -
- 3. (a) Explain the FIR filter design by Fourier series method.

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(b) Design a linear phase FIR low pass filter using hamming window by taking 5 samples of window sequence and with a cutoff frequency,  $w_c = 0.35\pi$  radians / sample. 7

### OR

- 4. (a) Compare rectangular and hamming window functions. 6
  - (b) Explain the procedure for designing FIR filters using windows. 7
- 5. (a) Why impulse invariant method is not preferred in the design of IIR filter other than low pass filter ? 4
  - (b) Consider a LTI system governed by the equation. y(n)+0.8301y(n-1)+0.7348y(n-2) = x(n-2). Discuss the effect of coefficient quantization on pole locations, when the coefficients are quantized by,
    - (i) 3-bits by truncation.
    - (ii) 4-bits by truncation.

#### OR

6. (a) For the analogue transfer function,

$$H(S) = \frac{0.8}{S^2 + 1.6S + 9.64}$$

Determine H(t) using bilinear transformation if (i) T = 1 sec. and (ii) T = 0.6 Sec. 7

(b) What is warping effect ? What is its effect on magnitude and phase response ?

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### SECTION B

- 7. (a) Explain the process of polyphase decomposition of an IIR filter with an example. 7
  - (b) Derive an expression for the spectrum of output signal of an interpolator.

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# OR

- 8. (a) What do you mean by multirate DSP ?
  - (b) A signal x(n) is sampled with a Sampling frequency of 50 KHz. Design a decimator that down samples this signal by a factor D = 5. The following specifications should be fulfilled by the filter used in the decimator: Passband = 0-4 KHz
    Passband ripple : 0.1 dB
    Transition zone : 4.5 KHz
    Stopband attenuation : 30 dB
    Sketch the block diagram of the decimator and calculate order of FIR equiripple filter that fulfills the specifications. 10
- 9. (a) Explain the role of DSP in Radar systems.
  - (b) Explain in detail about :---
    - (i) MAC unit.
    - (ii) Pipelining

Used in DSP processors.

#### OR

- 10. (a) What is the total memory space in TMS320C6713 DSP processor and how is it divided between program, data and I/O?
  - (b) Draw and explain the architecture of DSP processor TMS 320C6713. 9
- 11. (a) Explain various phases in sub-band coding algorithm.
  - (b) Prove the orthogonality relations of Doubenchies Wavelets.

### OR

- 12. (a) Prove "If f(t) has continuous wavelet transform given by  $CWT_f(a, b)$  then f'(t) = f(t-b') leads to the following transform.  $CWT_f(a, b) = CWT_f(a, b-b')$ ".
  - (b) Explain different steps to be followed for analysis of continuous wavelet transform. Show related signals also. 5



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