

M.E. First Semester (Electronics & Tele.) (Full Time (C.G.S.- New)  
**13334 : Digital Signal Processing and Applications : 1 ENTC 4**

P. Pages : 2

Time :



**AW - 3900**

Max. Marks : 80

- Notes :
1. Answer **three** question from Section A and **three** question from Section B.
  2. Due credit will be given to neatness and adequate dimensions.
  3. Assume suitable data wherever necessary.
  4. Use of pen Blue/Black ink/refill only for writing the answer book.

**SECTION – A**

1. a) What are the possible types of impulse for linear phase FIR filter? What are the conditions to be satisfied for constant phase delay in linear phase FIR filters. **7**  
b) State the advantage and disadvantage of digital filter. Compare it with analog filter. **6**

**OR**

2. a) Derive the expression to determine the poles of butterworth filter. **7**  
b) Describe Inverse Chebyshev Filter. **6**
3. a) Explain the procedure for designing FIR filter using windows. **7**  
b) Discuss equiripple linear phase FIR filter. **6**

**OR**

4. Design an ideal low pass filter whose desired frequency response **13**  
$$H_d(e^{j\omega}) = 1, \quad \frac{\pi}{3} \geq \omega \geq -\frac{\pi}{3}$$
$$= 0, \quad \pi \geq \omega \geq \frac{\pi}{3}$$
using Hamming window  
i) Determine the impulse response for  $N = 9$ .  
ii) Determine  $H(Z)$ .
5. Explain the design of IIR filter using Impulse Invariant Technique. What is pole mapping? **13**  
Give the steps to design digital filter using Impulse Invariant method.

**OR**

6. a) Determine  $H(Z)$  for a Butterworth filter satisfying the following specification **8**  
$$0.8 \leq |H(e^{j\omega})| \leq 1; \quad 0 \leq \omega \leq \frac{\pi}{4}$$
$$|H(e^{j\omega})| \leq 0.2; \quad \frac{\pi}{2} \leq \omega \leq \pi$$
Assume  $T = 0.1$  sec.  
Apply Bilinear transformation method.  
b) What is bilinear transformation. Give its properties, advantage and disadvantage. **5**

**SECTION – B**

7. a) Explain in detail all the identities of multirate DSP? 7
- b) Explain with block diagram the general polyphase framework for decimator and interpolator. 7

**OR**

8. a) The T.F. of an IIR filter is 7
- $$H(Z) = \frac{1 + 0.32Z^{-1} + 0.58Z^{-2}}{1 + 0.7Z^{-1} + 0.4Z^{-2}}$$
- perform polyphase decomposition of  $H(Z)$  to decompose into
- i) 2 – section
- ii) 4 – section
- b) What are the error in QMF filter bank? Explain how alias free QMF realization is achieved. 7
9. a) Explain different addressing modes of DSP Processor TMS 320 C 6713. 7
- b) Differentiate between Van – Neumann and Harvard Architecture. 7

**OR**

10. a) Explain the pipeline operations in DSP Processor TMS 320 C 6713. 7
- b) Write any 4 special instructions of TMS 320 C 67XX processor that are suitable for signal processing application and explain them. 7
11. a) Discuss briefly the various applications of wavelet transform. 6
- b) 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 7
- $$CWT_{f'}(a, b) = CWT_f(a, b - b') "$$

**OR**

12. a) Explain the various steps to compute continuous wavelet transform of a given signal. 7
- b) Prove the orthogonality relation of Daubechies wavelet. 6

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