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**Question Paper Code : X 60450**

B.E./B.Tech. DEGREE EXAMINATIONS, NOVEMBER/DECEMBER 2020  
Fifth Semester

Electronics and Communication Engineering  
EC 2302/EC 52– DIGITAL SIGNAL PROCESSING  
(Regulations 2008)

(Common to PTEC 2302 Digital Signal Processing for B.E. (Part-Time)  
Electronics and Communication Engineering – Fourth Semester – Regulations 2009)

Time : Three Hours

Maximum : 100 Marks

Answer ALL questions

PART – A

(10×2=20 Marks)

1. State the advantages of FFT over DFTs.
2. What is meant by bit reversal ?
3. Mention the advantages of cascade realization.
4. Convert the given analog transfer function  $H(s) = \frac{1}{s+a}$  into digital by impulse invariant method.
5. State the effect of having abrupt discontinuity in frequency response of FIR filters.
6. State Gibb's phenomenon.
7. What is scaling ?
8. What is dead band of a filter ?
9. Give the steps in multistage sampling rate converter design.
10. Write any four applications of multi-rate signal processing.



11. a) Explain in detail about overlap add method and overlap save method for filtering of long data sequences using DFT. (16)

(OR)

- b) Develop a 8 point DITFFT algorithm. Draw the signal flow graph. Determine the DFT of the following sequence.

$x(n) = \{1, 1, 1, 1, 0, 0, 0, 0\}$  using the signal flow graph. Show all the intermediate results on the signal flow graph. (16)

12. a) Discuss the steps in the design of IIR filter using Bilinear transformation for any one type of filter. (16)

(OR)

- b) Convert the following pole-zero IIR filter into a lattice ladder structure. (16)

$$H(z) = \frac{[1 + 2z^{-1} + 2z^{-2} + z^{-3}]}{\left[1 + \left(\frac{13}{24}\right)z^{-1} + \left(\frac{5}{8}\right)z^{-2} + \left(\frac{1}{3}\right)z^{-3}\right]}$$

13. a) i) Determine the frequency response of FIR filter defined by  $y(n) = 0.25x(n) + x(n-1) + 0.25x(n-2)$ .

Calculate the phase delay and group delay. (8)

- ii) Discuss the design procedure of FIR filter using frequency sampling method. (8)

(OR)

- b) Design a FIR filter with the following desired specification.

$$H_d(e^{j\omega}) = \begin{cases} 0, & -\frac{\pi}{4} \leq \omega \leq \frac{\pi}{4} \\ e^{-j2\omega}, & \frac{\pi}{4} \leq |\omega| \leq \pi \end{cases}$$

Using a Hanning window with  $N = 5$ . (16)



14. a) Explain the quantization process and errors introduced due to quantization. **(16)**

(OR)

b) i) Explain how reduction of product round-off error is achieved in digital filters. **(8)**

ii) Explain the effects of coefficient quantization in FIR filters. **(8)**

15. a) Explain with block diagram the general poly phase frame work for decimator and interpolator. **(16)**

(OR)

b) Implement a two stage decimator for the following specifications :

Sampling rate of the input signal = 20,000 Hz

$M = 100$

Passband = 0 to 40 Hz

Transition band = 40 to 50 Hz

Passband ripple = 0.01.

Stopband ripple = 0.002. **(16)**

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