Total No. of Questions : 10]

**SEAT No. :** [Total No. of Pages : 4

**P2953** 

## [5669]-542

**T.E.** (**E** & **TC**) **DIGITAL SIGNAL PROCESSING** 

2015 Pattern)

*Time : 2<sup>1</sup>/<sub>2</sub> Hours*]

[Max. Marks : 70

Instructions to the condidates:

- 1) Attempt Q.1 or Q.2, & Q.3 or Q.4, & Q.5 or Q.6, & Q.7 or Q.8, & Q.9 or Q.10.
- Draw suitable diagrams where necessary. 2)
- Figures to the right indicate full marks. 3)

Show relationship between analog frequencies and digital frequencies.[4] *Q1*) a)

- An analog signal contains frequencies upto 10kH2 b) [6] What is range of sampling frequencies must be selected for faithful
  - reconstruction of signal?
  - If signal is sampled with sampling frequency 8kHz what is folding ii) frequency?
  - Does aliasing occours if ip is 5kHz? if yes calculate alised iii) frequencies from original frequencies

(Consider sampling freq to be 8k)

Does aliasing occours at i/p frequency if i/p is 9kHz? if yes calculate aliased.

OR

- Q2) a) Consider signal  $x(t) = 5 \sin (500 \text{ mt})$  if signal is sampled at Fs = 1500 Hz[4]3
  - What is DT signal obtained after sampling? i)
  - ii) Find frequency of DT signal.
  - Find DT signal for sampling frequency Fs iii)
  - State & Prove following properties in Z transform b) ).200.200.200 2.200.200 2.200.200

[6]

- Scaling property. i)
- ii) Time shift property.

*P.T.O.* 

Q3) a) Determine IZT of following functions

(23) a) Determine IZT of following functions [8]  
a) 
$$X(z) = \frac{(8z-19)}{(z-2)(z-3)}$$
  
 $x(n)$  is causal  
b) Obtain z transform of following  
 $x(n) = (0.5)^n u(n) + (-0.2)^n u(n-3)$   
OR  
(24) a) Compute IDFT by matrix method  
 $X (k) = (10, -2 + 2j, -2, -2 - 2j)$   
b) Compute 8 point DFT of sequence  
 $x(n) = (0, 12, 3)$  & draw magnitude & phase plot  
(6)  
 $x(n) = (0, 12, 3)$  & draw magnitude & phase plot  
(5) a) Compare IIR & Fir filters or following points. [6]  
b) Filter governing mathematical equation  
ii) Memory requirment  
iii) Stability  
iv) Recursiveness  
v) Phase response  
v) Phase response  
v) Phase response  
v) Phase response  
(4)  
(5) Convert analog filter with system function It(s) into drata IIR filter using impulse invariance method  
 $H(s) = \frac{10}{s^2 + 75 + 10}$ 

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OR Compare butterworth filter & chebyshev filter on following points. [4] **Q6**) a) Frequency response i) Order for given specification ii) Transition band iii) Phase response & pole location iv) Design Butter worth filter for following specification. b) [8]  $0 \le F \le 1000 Hz$  $0.8 \leq |H|$  $(S) \le 1$  $F \ge 5000 Hz$ (s) |≤0.2 Draw direct form 2 realization for the following [4] c) Q7) a) Explain following window functions [6] Rectangular window i) Hamming window ii) Hanning window iii) Design linear phase FIR filter using hamming window with cutoff freq. b) 0.2 rad/sec and 0.3 rad/se use M = 7Calculate  $1^{st}$  two filter coefficients only (n=0 & n=1) OR What is gibbs phenomenon? How the effect of gibbs phenomenon is (0.8) a) reduced? [4] Using frequency sampling method design law pass FIR filter to meet b) following specification. Pass band - 0-5kHz Filter length = 9 Sampling frequency  $\rightarrow$  18kHz Obtain h(n) for n = 0.1 only [12]

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**Q9**) a) Write a note on digital cross over audio system. [6] With the help of block diagram explain enhancement of ECG signal for b) heart rate detection. [6] Explain speech compression & decompression with block diagram. [6] c) OR *Q10*)a) Explain compact disc recording system. [6] Compare digital signal processing & analog s/g processing on following b) points Accuracy & component tolerance Cost Upgradation adaptation Implementation Repeatability Versatility [6] Explain how the defective gear tooth can be identified using vibration c) analysis? [6] Ano. 200 aning of the static