

T.E. (Computer) (Semester – I) Examination, 2010 DIGITAL SIGNAL PROCESSING (New) (2008 Course)

	(2008 Course)	nadola kiki i
Time: 3 H	lours	Max. Marks: 100
Instru	and Q. 7 or Q. 8, Q. 9 or Q. 10, Q. 11 2) Answers to the two Sections should be books. 3) Neat diagrams must be drawn where 4) Black figures to the right indicate full 5) Use of logarithmic tables, slide rule, pocket calculator and steam tables is 6) Assume suitable data, if necessary.	or Q. 12 from Section II. we written in separate ever necessary. Il marks. Mollier charts, electronic s allowed.
	SECTION – I	
	ow an arbitrary DT sequence can be represented by obtain linear convolution of $u(n)$ and δ (n) using graph	•
B) Sta	ate linearity and time invariant properties of a DT s	ystem and test it for
y(r	(n) = x(n) - 2.x (n - 2). OR	aftengregue entrett.(A q a cereb bass (X)H
	ow a DT system is represented/described in time dom sponse of a DT system and show that for a causal	
h(r	n) = 0 for n < 0	
	Thy the problem of aliasing is observed during sar at all the frequencies $F_K = F_O \pm K.F_S$, K: integer are	mpling process ? Show the aliases of frequency
	ompare N point DFT with FT. What is the signification is necessary to have $N \ge L$ where L: length of a D'	
B) Sta	ate and prove windowing theorem of a Fourier Tra	nsform (FT).

OR



4. A) Obtain DT frequencies {W_K} for 6 point DFT. State the convolution and periodicity property of DFT. What is DFT spectral leakage?

10

B) Explain how DFT can be used for linear filtering. How N-pt. circular convolution can be used to obtain linear convolution?

8

5. A) How computational complexity of N-point DFT is reduced using FFT algorithm? What is in-place computation and bit-reversal indexing in FFT?

8

B) Using ZT properties, obtain ZT of a DT signal x(n) = 2n. u(-n-1) sketch the ROC.

8

OR

6. A) Derive the first stage of DIF (decimation in frequency) FFT algorithm. Draw the basic butterfly structure for the same.

8

B) Obtain a causal sequence x(n) from its ZT

8

$$X(Z) = \frac{3}{Z - \frac{1}{4} - \frac{1}{8Z}}$$

SECTION - II

7. A) Define a system function H(Z). What is pole zero plot of a system? Determine H(Z) and draw a pole zero plot for a system.

10

$$y(n) + \frac{3}{4}y(n-1) + \frac{1}{8}y(n-2) = x(n) + x(n-1)$$

B) Express system function H(Z) for FIR and IIR system from the Nth order difference equation. How properties of DT system can be described using H(Z)?

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quencies F = F + K F K meeters no

8. A) Knowing H(Z), how frequency response of a system can be obtained? Use simple geometric construction to obtain the frequency response of a system having $h(n) = \{0.5, 0.5\}.$

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B) Determine the impulse response of a system.

8

$$y(n) = 2x(n) - x(n-1) - 3y(n-1) - 2y(n-2)$$



9. A) Define a DT filter. What do you mean by a linear phase response? What is group delay? State the advantages and disadvantages of FIR filter over an IIR filter.

8

B) What is frequency prewarping? Design second order low pass filter using BLT method having

 $H(S) = \frac{1}{S^2 + \sqrt{2} S + 1}$ with cut-off freq. $F_c = 1$ KHz and sampling frequency

 $F_s = 10$ KHz. Use frequency prewarping.

OR

10. A) Compare Hanning window with rectangular window. Write the algorithmic steps to design an FIR filter using Hanning window.

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8

B) Stable analog filter always gives stable digital filter. Justify the statement. Explain impulse-invariance method for the design of an IIR filter.

8

11. A) Obtain direct and cascade form FIR filter structure for a system having

 $H(Z) = (1 + Z^{-1}) (1 + \frac{1}{2} Z^{-1} + \frac{1}{2} Z^{-2} + Z^{-3})$

8

B) Obtain direct form - I and direct form - II IIR filter structure for a system

 $H(Z) = \frac{Z^2 + \frac{1}{3}Z}{Z^2 - \frac{3}{4}Z + \frac{1}{8}}$

8

OR

12. A) How DSP processors are different than conventional microprocessors? List the important features of ADSP 21XX DSP processor. What is DAG?

8

B) How image is represented by digital computer? Explain the application of DSP in image processing w.r.t. image enhancement.

8