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## Question Paper Code:1065092

## B.E. / B.Tech. DEGREE EXAMINATIONS, NOV/ DEC 2024 Fifth Semester Electronics and Communication Engineering U20EC502 – DIGITAL SIGNAL PROCESSING (Regulation 2020)

Γime:	Three Hours	Maximum:	100	Marks

Answer ALL questions

 $PART - A \qquad (10 \times 2 = 20 \text{ Marks})$ 

- 1. Recall the analysis and synthesis equation of DFT.
- 2. Compare overlap save method and overlap add method.
- 3. State Gibbs phenomenon.
- 4. Specify how the ISR filter can be designed?
- 5. Label the quantization noise model for a first order system.
- 6. Infer the dead band of the filter.
- 7. State the basic operations of Multirate signal processing.
- 8. Point out the need for anti imaging filter after up sampling a signal?
- 9. Brief the features of MAC unit.
- 10. Recall the applications of Digital Signal Processor.

11.(a) (i) State and Prove the following properties of DFT.

(6)

(12)

- (1) Time reversal (2) Parsavel's Theorem
- (ii)Compute the 8-Point DFT of the sequence  $x(n)=\{1,2,3,4,4,3,2,1\}$  using radix-2 DIT FFT Algorithm. (10)

(OR)

- (b) Find the output y(n) of a filter, the input is given by  $x(n) = \{3, -1, 0, 1, 3, 2, 0, 1, 2, 1\}$  and  $h(n) = \{1, 1, 1\}$  using overlap save method and overlap add method. (16)
- 12.(a) (i) The desired frequency response of a digital filter is.

 $H_d(e^{j\omega}) = e^{-j3\omega}$ ;  $-\pi/4 \le \omega \le \pi/4$ 0;  $\pi/4 \le \omega \le \pi$ 

Determine the filter co-efficient if the window function is

$$\omega(n) = 1 ; 0 \le \omega \le 5$$
  
0; otherwise

(ii) Draw the direct form realization of  $H(Z)=1+2z^{-1}-3z^{-2}-4z^{-3}+5z^{-4}$  (4)

(OR)

(b) Design a digital butterworth LPF which satisfies the constraints using bilinear transformation with T=1 sec. (16)

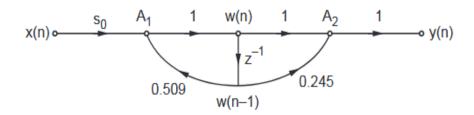
$$0.707 \le |H(\omega)| \le 1.0$$
;  $0 \le \omega \le \pi/2$   
 $|H(\omega)| \le 0.2$ ;  $3\pi/4 \le \omega \le \pi$ 

13.(a) i) Derive the expression for Quantization noise power.

- (6)
- ii) Find the steady state variance of the noise in the output due to quantization of input for the first order filter y(n) = ay(n-1) + x(n). (10)

(OR)

(b) For the digital network shown in figure Find H (z) and Scaling factor. (16)



14.(a)	<ul><li>i) Show that up sampler and down sampler are time variant systems.</li><li>ii) Consider the signal x(n)=nu(n)</li></ul>	
	1) Determine the spectrum of the signal.	(4)
	2) The signal is applied to a decimator that reduces the sampling rate by a factor	
		(12)
		,
	(OR)	
(b)	i)Describe and derive sampling rate conversion by a rational factor I/D in Multissignal processing.	rate (10)
	ii) Outline the concept of Multistage implementation of sampling rate conversion.	. ,
15.(a)	<ul><li>i) Enumerate the various features of Von Neumann and Harvard architectures.</li><li>ii) Illustrate about different stages of pipelining and specify its importance.</li></ul>	(8) (8)
	(OR)	
(b)	With relevant diagrams, determine the detail about the basic architecture of figure point processors TMS320C5X.	xed (16)
	XXX	