

Code No: RT32042

R13

SET - 1

III B. Tech II Semester Regular/Supplementary Examinations, April - 2017

DIGITAL SIGNAL PROCESSING

(Electronics and Communication Engineering)

Time: 3 hours

Maximum Marks: 70

Note: 1. Question Paper consists of two parts (**Part-A** and **Part-B**)

2. Answering the question in **Part-A** is compulsory

3. Answer any **THREE** Questions from **Part-B**

PART -A

- 1 a) Test whether the following signal is periodic or not ,if periodic find the fundamental period $\sin \sqrt{2} \pi t$ [4M]
- b) Find the DFT of a sequence $x(n) = \{1, 1, 2, 2\}$ [4M]
- c) Give block diagram representation of linear constant-coefficient difference equations. [4M]
- d) By impulse invariant method obtain the digital filter transfer function and the differential equation of the analog filter $h(s) = 1/s+1$ [4M]
- e) What are the applications of multi rate DSP? [3M]
- f) List special feature of DSP architecture. [3M]

PART -B

- 2 a) Determine whether each of the following systems defined below is (i) casual (ii) linear (iii) dynamic (iv) time invariant [12M]
 - (i) $y(n) = \log_{10}[\{x(n)\}]$
 - (ii) $y(n) = x(-n-2)$
 - (iii) $y(n) = \cosh[nx(n) + x(n-1)]$
- b) Give the frequency domain representation of discrete time signals. [4M]
- 3 a) Compute the DFT for the sequence $\{1, 2, 0, 0, 0, 2, 1, 1\}$. Using radix -2 DIF FFT and radix -2 DIT- FFT algorithm. [8M]
- b) Derive the equation to implement a butterfly structure In DITFFT algorithm. [8M]
- 4 a) Realize the filter $H(z) = (z^{-1} - a)(z^{-1} - b) / (1 - az^{-1})(1 - bz^{-1})$ in cascade and parallel forms. [8M]
- b) State and prove time convolution property of Z-Transforms. [8M]
- 5 a) Obtain the impulse response of digital filter to correspond to an analog filter with impulse response $h_a(t) = 0.5 e^{-2t}$ and with a sampling rate of 1.0kHz using impulse invariant method. [8M]
- b) Compare bilinear transformation and impulse invariant mapping. [8M]
- 6 a) Explain the decimation and interpolation process with an example. Also find the spectrum. [8M]
- b) The sequence $x(n) = [0, 2, 4, 6, 8]$ is interpolated using interpolation sequence $b_k = [1/2, 1, 1/2]$ and the interpolation factor is 2. find the interpolated sequence $y(m)$. [8M]
- 7 a) Describe the multiplier/adder unit of TMS320c54xx processor with a neat block diagram. [8M]
- b) What are interrupts? What are the classes of interrupts available in the TMS320C5xx processor? [8M]

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2. Answering the question in **Part-A** is compulsory

3. Answer any **THREE** Questions from **Part-B**

PART -A

- 1 a) Test whether the following signal is periodic or not, if periodic find the fundamental period $\sin 20\pi t + \sin 5\pi t$ [4M]
- b) Find the values of WN_k , When $N=8$, $k=2$ and also for $k=3$. [4M]
- c) Draw the direct form realization of FIR system. [4M]
- d) What are the properties of chebyshev filter? [3M]
- e) Find the spectrum of exponential signal decimated by factor 2. [4M]
- f) What are the advantages of VLIW architecture? [3M]

PART -B

- 2 a) Determine the impulse response of the filter defined by $y(n)=x(n)+by(n-1)$. [8M]
- b) A system has unit sample response $h(n)$ given by $h(n)=-1/\delta(n+1)+1/2\delta(n)-1/4\delta(n-1)$. Is the system BIBO stable? Is the filter causal? Justify your answer. [8M]
- 3 a) Find the DFT of the sequence $x[n]=\{1,2,3,4,5,6,7,8\}$. [8M]
- b) Explain the use of FFT algorithms in linear filtering and correlation. [8M]
- 4 a) Determine the cascade and parallel realization for the system transfer function $H(z) = 3(z^2+5z+4) / (2z+1)(z+2)$. [8M]
- b) State and prove frequency convolution property of Z-Transforms. [8M]
- 5 a) Design an ideal high pass filter with a frequency response $H_d(e^{j\omega}) = 1$ for $\pi/4 \leq |\omega| \leq \pi$
 $= 0$ for $|\omega| \leq \pi/4$ Find the values of $h(n)$ for $N = 11$ using Hamming window. Find $H(z)$ and determine the magnitude response. [8M]
- b) Derive the expression for Bi linear Transform. [8M]
- 6 a) Explain the operation used in DSP to increase the sampling rate. [8M]
- b) The sequence $x(n)=[0,2,4,6,8]$ is interpolated using interpolation sequence $b_k=[1/2,1,1/2]$ and the interpolation factor is 2. find the interpolated sequence $y(m)$. [8M]
- 7 a) Explain the different types of interrupts in TMS320C54xx Processors. [8M]
- b) Describe any four data addressing modes of TMS320c54xx processor. [8M]

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Time: 3 hours

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- Note: 1. Question Paper consists of two parts (**Part-A** and **Part-B**)
 2. Answering the question in **Part-A** is compulsory
 3. Answer any **THREE** Questions from **Part-B**

PART -A

- 1 a) Test the following systems for time invariance $y(n)=n x^2(n)$ [4M]
- b) Define DFT and IDFT [4M]
- c) What are the applications of Z-Transforms? [4M]
- d) What are the advantages of Kaiser widow? [4M]
- e) What are "decimation" , "decimation factor "and "down sampling"? [3M]
- f) List the on-chip peripherals [3M]

PART -B

- 2 a) Determine and sketch the magnitude and phase response of the following systems [12M]
 - (i) $y(n) = 1/3 [x(n) + x(n-1) + x(n-2)]$
 - (ii) $y(n) = 1/2[x(n) - x(n-1)]$ (iii) $y(n) - 1/2y(n-1)=x(n)$
- b) Determine the impulse response of the filter defined by $y(n)=x(n)+by(n-1)$. [4M]
- 3 a) Determine IDFT of the following [8M]
 - (i) $X(k)=\{1,1-j2,-1,1+j2\}$ (ii) $X(k)=\{1,0,1,0\}$
- b) Find the DFT of the sequence $x[n]=\{1,2,3,4,5,6,7,8\}$ using DITFFT. [8M]
- 4 a) Obtain the direct form I, direct form II and Cascade form realization of the [8M]

following system functions.

$$Y(n) = 0.1 y(n-1) + 0.2 y(n-2) + 3x(n) + 3.6 x(n-1) + 0.6 x(n-2).$$
- b) Explain Transposed forms. [8M]
- 5 a) Comparison of FIR and IIR filters. [8M]
- b) What is Hamming Window function? Obtain its frequency domain characteristics. [8M]
- 6 a) What is Multi Rate Signal Processing? Explain any two applications of multirate [8M]

signal processing.
- b) Derive the Frequency domain Transfer function of a Decimator. [8M]
- 7 a) List the major architectural features used in DSP system to achieve high speed [8M]

program execution.
- b) With examples explain the different addressing formats supported by DSP [8M]

processors for various signal processing applications.

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 2. Answering the question in **Part-A** is compulsory

 3. Answer any **THREE** Questions from **Part-B**

PART -A

- 1 a) Test the following systems for time invariance $a^{x(n)}$. [3M]
- b) What are the advantages of FFT over DFT. [4M]
- c) Find the Z-transform of $x(n) = (1/8)^n u(n)$ and its ROC. [4M]
- d) What is the necessary and sufficient condition for linear phase Characteristics in FIR filter? [4M]
- e) Explain the term up sampling and down sampling. [3M]
- f) What are the different stages in pipelining? [4M]

PART -B

- 2 a) A system has unit sample response $h(n)$ given by $h(n) = -1/\delta(n+1) + 1/2\delta(n) - 1/4\delta(n-1)$. Is the system BIBO stable? Is the filter causal? Justify your answer [8M]
- b) Give the frequency domain representation of discrete time signals and systems. [8M]
- 3 a) How is the FFT algorithm applied to determine inverse discrete Fourier transform? [8M]
- b) Derive the equation to implement a butterfly structure In DIFFFT algorithm [8M]
- 4 a) Obtain the direct form I, direct form II and Cascade form realization of the following system functions. [8M]
 $Y(n) = 0.1 y(n-1) + 0.2 y(n-2) + 3x(n) + 3.6 x(n-1) + 0.6 x(n-2)$.
- b) Prove that FIR filter has linear phase if the unit impulse response satisfies the condition $h(n) = h(N-1-n)$, $n=0,1,\dots,M-1$. Also discuss symmetric and antisymmetric cases of FIR filter. [8M]
- 5 a) Determine $H(Z)$ for a Butterworth filter satisfying the following specifications: [8M]
 $0.8 \leq |H(e^{j\omega})| \leq 1$, for $0 \leq \omega \leq \pi/4$
 $|H(e^{j\omega})| \leq 0.2$, for $\pi/2 \leq \omega \leq \pi$
 Assume $T = 0.1$ sec. Apply bilinear transformation method
- b) Use bilinear transformation method to obtain $H(Z)$ if $T = 1$ sec and $H(s)$ is $1/(s+1)(s+2)$, $1/(s^2 + \sqrt{2}s + 1)$. [8M]
- 6 a) With necessary derivation explain the operation of sampling rate conversion by a non integer. [8M]
- b) The sequence $x(n) = [0, 3, 6, 9]$ is interpolated using interpolation sequence $b_k = [1/3, 2/3, 1, 2/3, 1/3]$ and the interpolation factor of 3. Find the interpolated sequence $y(m)$. [8M]
- 7 a) Explain Memory Access schemes in DSPs. [8M]
- b) Explain the memory interface block diagram for the TMS 320 C5x processor. [8M]
