## BITS, PILANI – DUBAI CAMPUS

International Academic City, Dubai Year IV – Semester II 2012 – 2013

Comprehensive examination

Course No.: EEE C 415

Course Title: DSP

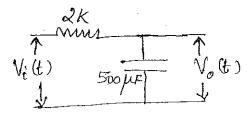
Date: June 03, 2013

Time: 3 Hrs...

Max. Marks = 40

(Any assumptions made should be indicated clearly)

1. Using Bilinear – Z transform method of designing digital filters, determine the transfer function and hence the difference equation for the digital equivalent of the analog RC filter shown in figure below. Assume a sampling frequency of 100 Hz and a cut off frequency of 30 Hz. (4M)



- 2. a) Distinguish between the frequency response of Chebyshev type I filter for N is odd and even (1M)
  - b) Determine the frequency response of an IIR filter characterized by

y (n) = 0.3x (n) - x (n-1) + 0.5x (n-2) - y(n-1) Compute the phase delay and group delay of the filter

(6M)

- 3. Derive the impulse response coefficients  $h_D(n)$  of an ideal band pass filter (3M)
- 4. A 4-point linear phase FIR filter is characterized by the following frequency samples. (4M)

$$|H(k)| = 1, k = 0$$
  
= ½, k = 1,3  
= 0. k = 2

- a) Determine the four coefficients of the filter which could be implemented.
- b) Determine the transfer function of the Recursive FIR frequency sampling filter.
- With a neat block diagram explain the principle of Digital signal processing as applied to any specific application of your choice. (3M)
  - b) Consider an audio band signal sampled at a rate of 100 KHz. It is required to down rate the sampling frequency to 12 KHz. The highest frequency of interest after decimation is 4KHz. Design a suitable optimum frequency converter which will satisfy the following overall specifications.

Pass band ripple = 0.1;

Stop band ripple = 0.01

Filter length N =  $\frac{-10 \log (\delta_s \delta_p) - 13}{14.6 \Delta f}$  + 1; where  $\Delta f$  is the normalized frequency.

Draw also the frequency response of the designed decimator stages.

(5M)

О.			$7 z^{-1} + z^{-2}$ ) / (1-			fit.		
	The	filter is to b	e implemente	d using an 8b	it system. Assu	ıming that a s	econd order	· canonic
	secti	on is used to	realize the filt	er, and each p	roduct has to b	e quantized to	8 bit,	
a) b)			ation diagram s <sub>2</sub> - norm scaling	_			3+3M)	
7.	Star	ting with the	e equation for t	he mean squa	re error, derive	the Wiener –	Hopf equation	on to
	estir	mate the op	timum weights	of the adaptiv	e filter.	. (	3M)	
8.	With	reference to	o TMS320C5X p	processor, choo	ose the appropi	riate answer	(1x5M)	- Aller
	i)	The symb	ool used to indi	cate the imme	diate address n	node of operar	nds is	,
		a) \$	b) *	√c) #	d) &	e) %		
	ii)	a) Instruc c) Use pip	tion Cache 🗸	b) Number o	f functional Uni	ts	ollowing aspe	ect?
	iii)	The reg	gister in which bit wide.	the the result	of multiplication	on is stored is		and its
	"محرة	a) PREG, 3	32 b) PREG	i, 16 c) TREGO,	16 d) TREGO,	, 32		
	iv)	a) indirect	essing mode that addressing rsed addressing	b) circula	mode	utation is		
	v)	40C5H , 2		300H respectiv	vely initially. W		ilter (3+3M)  ner – Hopf equation to (3M)  ver (1x5M)  perands is the following aspect?	
	V	a) 2400H	b) 2300H	c) ·	40C5H d)	0003Н		
			****	******	******			

#### BITS PILANI, DUBAI CAMPUS II SEMESTER 2012 - 2013

Course Code: EEE C415

Course Title: Digital Signal Processing

TEST-2 - OB

Date: 01.05.2013 Max.Marks:30

Weightage:15%

Duration: 50 minutes

(All questions carry 10 Marks each)

1. Determine the FIR filter coefficients h(n) of a filter with the following frequency response by frequency sampling method for N= 7  $H_d(e^{jw}) = e^{-j(N-1)\omega 2}$ ;  $0 \le |\omega| \le \pi/2$ 

10

$$e^{-3(\pi^2-1)\omega 2}$$
;  $0 \le |\omega| \le \pi/2$   
=  $0$   $\pi/2 \le |\omega| \le \pi$ 

2. Compare the single stage, two stage, three stage and multistage realization of the decimator with the following specifications. Input sampling frequency is 10KHz, output sampling frequency is 500 Hz. The

decimation filter H(z) has the pass band edge frequency 150 Hz, stop band edge frequency 180 Hz, pass band ripple 0.002 and stop band ripple 0.001.

3. Following is a code segment of TMS 320 C50 program. Comment each code line with its operation and write the function the program executes.

ADDRESS1 IN XN, ADC ZAC LT XNM2 MPY H2 LTD XNM1 MPY H1LTD XN MPY H0 **APAC** SACH YN,1 OUT YN, DAC В ADDRESS1

#### BITS PILANI DUBAI CAMPUS

Dubai International academic City, Dubai Year IV – Semester II 2012–2013 Test I (Closed Book)

Course No.: EEE C 415

Course Title: DSP

Date: March 13, 2013

Time: 50 Minutes

Max. Marks = 30

( Assume any data missing and clearly state the assumptions made)

1. Specify and sketch a suitable pole-zero diagram of a simple low pass discrete-time filter with the following specifications. (9M)

Cut-off frequency: 1KHz, Sampling frequency: 10KHz

Obtain the transfer function of the filter from the pole-zero diagram. Determine the amplitude and phase response at 1KHz, 2.5 KHz and 5KHz. Sketch the amplitude response of the filter.

- 2. The transfer function of a DT system has poles at z = 0.5, z = 0.1 + j 0.2 and zeroes at z = -0.25 and at z = 0.75.
  - a) Derive the system transfer function H(z).
  - b) Develop the difference equation
  - c) Draw the canonic form of realization of the system.

(8M)

- 3. Derive the impulse response coefficients  $h_D(n)$  of an ideal band pass filter (5M)
- 4. Obtain the coefficients of an FIR low pass digital filter to meet the specifications below using window method.

Pass-band edge frequency: 3.2 kHz

stop band attenuation = 40 dB

Transition width: 0.6KHz Sampling frequency: 8 KHz

Give your comments on the window used and the reason for your choice.

(8M)

MS.

# BITS, PILANI – DUBAI

DIAC, Dubai

Year IV – Semester II 2012–2013

Quiz II

Course	No.:	EEE	C 415

Course Title: DSP

Date: May 14, 2013

Time: 20 Minutes

Max. Marks = 10

(Question nos. 1-4 carries one mark each and Question Nos 5-7 carries 2 marks each)

	•						,	
			PART A	<u>1</u>				
1		g address of the da	ta memory page po	ointed by the	e DP register	r whose c	ontent is 7	
2		memory used wi vords long.	th C5X processors	s is split in	to	p	ages each	of
	∟a) 512,	128 b)	256,256 the rug phepares and a termines whether	c) 128,512	2	d) 1024,	54	
3.	0,1,4 or righ	ı <del>t shifted by 6 wi</del> _	th sign extension l	before it is TC	transferred / 	added to	the ACC	is
	a) CNF	b) INTM	c) HM	d) XF	e) PM	(1)	how di	Lather
4.	When MACI		ecuted, the produc				=	
		(Assu	PART B me the data missi	ng if any )				
Fo	Write the con		s before execution at registers which v				or each of	
	DP = 6	300h] = 04h;	[310h] = 24h;	[08]	[00H] = 03h			
	TREG0 = 221	ı	ACC = 765432	10h ARO	CR = 2530h		-	
	ARP = 7 A	AR7 = 2350h	INDX = 10h	[235	50h] = 132h	[50	h] = 4680h	
5.	LAMM * 0 +,	AR4 ->	ARP=4.	ART:	23604 00004681			EK+
ó.	NEG -> :	2 < ACC . B	9ABCDF0	)				50
7.	ADD 10h, 2	24	ーフ 090 h 32·10·	(shifte	d) 60	10 0/2	100 	÷
			165432 A O	4				

# BITS, PILANI – DUBAI

DIAC, Dubai Year IV – Semester II 2012–2013 Quiz II

Course No.: EEE C 415

Course Title: DSP

Date: May 14, 2013

Time: 20 Minutes

Max. Marks = 10

(Question nos. 1-4 carries one mark each and Question Nos 5-7 carries 2 marks each)

#### PART A

1		ing address		memory page	pointed b	y the DP r	egister	whose o	content is	7
2		ta memory _words lor		C5X process	ors is spl	it into	<del>.</del>	I	oages each	ı of
				56,256				l) 1024,		
- 3.	The statu 0,1,4 or r	s register l <del>ight shifte</del> r	Co oit that <del>det</del> I by 6 with	Lipey L. ernnines wheth sign extensio	n before i	R365 ( Her's 32-b t is transfe	it prod	J-age uct is l added t	in Stan eft shifted o the ACC	aller by rug sley is
	a) Cl	NF I	) INTM	c) HM	d) XF	e)	) PM	(チ)	home of	There.
4.	When MA		ction is exe	cuted, the prod	luct regist	er is shifted	d by the	e count	specified b	py
			(Assum	PART B ne the data mi	-	ny)	•			
F		content of t	he relevant	before execution registers which					for each of	f
	DP = 6			[310h] = 24	h;	[08F00H]	= 03h			
	TREG0 = 3	22h		ACC = 7654	43210h	ARCR = 2	2530h		•	
	ARP = 7	AR7 = 2	350h	INDX = 10	1	[2350h] =	132h	[5	[0h] = 4680	Oh
5.	LAMM * (	) +, AR4								
6.	NEG									
7.	ADD 10h,	2								

## BITS PILANI DUBAI CAMPUS

DIAC, Dubai Year IV – Semester II 2012–2013 Ouiz I

Course No.: EEE C 415

Course Title: **DSP** 

Date: March 05, 2013

Time: 20 Minutes

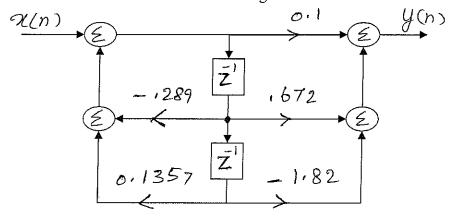
Max. Marks = 10

#### (All questions carry 2 marks each)

1. Match the following with most appropriate connectives

1.	Butterworth Filter	Minimum no. of coefficients ( )
2.	IIR filter	Maximally Flat frequency response ( )
3.	FIR filter	Unstable system ( )
4.	Bilinear Z-transformation	Recursive filter ( )
		Feedback filter ( )
		Linear Phase Characteristics ( )
		Pole zero mapping to discrete domain ( )
		Poles at maximum and zeroes at minimum
		response ( )

2. Write the transfer function of the digital IIR filter structure shown below.



- 3. Derive the amplitude response of the digital filter defined by y(n) = 0.25 x(n) + 0.4y(n-1) 0.25 y(n-2).
- 4. Sketch the pole -zero diagram of the discrete time system whose transfer function is given by  $H(z) = (1 1.6 z^{-1} + z^{-2}) / (1 1.5 z^{-1} + 0.8 z^{-2})$
- 5. Obtain the digital filter transfer function of the analog filter with H(s) = 0.3 / (s 2), by Impulse Invariant method.

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## BITS PILANI DUBAI CAMPUS

DIAC, Dubai Year IV – Semester II 2012–2013 Quiz I MS

Course No.: EEE C 415

Course Title: DSP

Date: March 05, 2013

Time: 20 Minutes

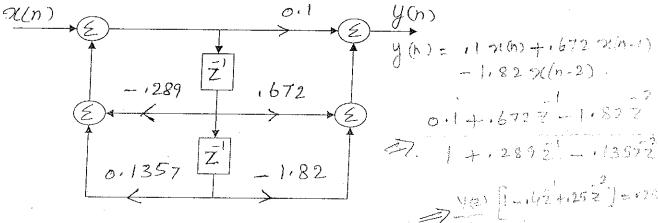
Max. Marks = 10

(All questions carry 2 marks each)

1. Match the following with most appropriate connectives

1.	Butterworth Filter	Minimum no. of coefficients ( )
2.	IIR filter	Maximally Flat frequency response ( 1 )
3.	FIR filter	Unstable system ( )
4.	Bilinear Z-transformation	Recursive filter (2)
		Feedback filter (2)
		Linear Phase Characteristics (3)
	·	Pole zero mapping to discrete domain ( 4 )
		Poles at maximum and zeroes at minimum
		response ( )

2. Write the transfer function of the digital IIR filter structure shown below.



- 3. Derive the amplitude response of the digital filter defined by y(n) = 0.25 x(n) + 0.4y(n-1) 0.25 y(n-2).
- 4. Sketch the pole -zero diagram of the discrete time system whose transfer function is given by  $H(z) = (1 1.6 z^{-1} + z^{-2}) / (1 1.5 z^{-1} + 0.8 z^{-2})$
- 5. Obtain the digital filter transfer function of the analog filter with H(s) = 0.3 / (s 2), by Impulse Invariant method.

$$h(t) = 0.3.e$$

$$H(z) = \frac{0.3}{1-e}$$

3.

H(Z): 0.25
$$1-0.4 = \frac{25.1}{1-0.4} = \frac{35.1}{1-0.4} = \frac{35.1}{1-0.4} = \frac{35.1}{1-0.25} = \frac{35.1}{1-0.25} = \frac{35.1}{1-0.25} = \frac{35.1}{1-0.4} = \frac{35.1}{1-0.25} = \frac{35.1}{1-0.25} = \frac{35.1}{1-0.25} = \frac{35.1}{1-0.25} = \frac{35.1}{1-0.25} = \frac{35.1}{1-0.4} = \frac{35.1}{1-0.25} =$$

Asna - 8½ Thaitamyr - 4½ Ishani - 3½ Ninad - 2½