



Seat Number

King Mongkut's University of Technology Thonburi
Final Examination
Semester 1 – Academic Year 2014

Subject: EIE 460 Digital Signal Processing

For: Electrical Communication and Electronic Engineering, 4th Yr (Inter. Program)

Exam Date: Monday, December 1, 2014

Time: 9.00-12.00 am.

Instructions:-

1. This exam consists of 7 problems with a total of 8 pages, including the cover.
2. This exam is opened book.
3. Answer each problem on the exam itself.
4. A calculator compiling with the university rule is allowed.
5. A dictionary is **not** allowed.
6. Do not bring any exam papers outside the exam room.

Remarks:-

- Raise your hand when you finish the exam to ask for a permission to leave the exam room.
- Students who fail to follow the exam instruction might eventually result in a failure of the class or may receive the highest punishment with university rules.

Exam No.	1	2	3	4	5	6	7	Total
Full Score	25	20	10	25	25	5	10	120
Graded Score								

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This examination is designed by
Dr. Raungrong Suleessathira; Tel: 9060

This examination has been approved by the committees of the ENE department.

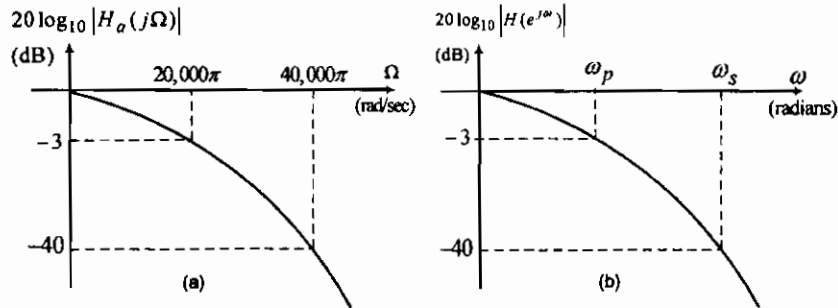
(Asst. Prof. Suwat Pattaramalai, Ph.D.)

Acting Head of Electronic and Telecommunication Engineering Department

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Student ID _____

1. The frequency response of an analog filter is shown in Figure (a). The sampling rate is $F_s = 50,000$ samples/sec. Use the FIR filter design. (25 marks)



- Find ω_p and ω_s .
- Find N by using Hamming window.
- Find the impulse response $h[n]$.
- If the magnitude of the frequency response of c. obtains -3 dB at a frequency less than ω_p and -40 dB at a frequency less than ω_s , how to improve the design to satisfy the requirement as shown in Figure (b)?
- Find the output signal of the filter that has frequency response as shown in Figure (b) if the input signal is $x[n] = \sin(\omega_p n)$.

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2. The specification of a digital filter is as follows. (20 marks)

$$|H(e^{j\omega})| \leq 0.01 \quad |\omega| \leq 0.25\pi$$

$$0.95 \leq |H(e^{j\omega})| \leq 1.05 \quad 0.3\pi < |\omega| \leq \pi$$

Use Kaiser window.

- Find A_s , ω_c , N and β .
- Use functions in Matlab to find the frequency response $H(e^{j\omega})$ where ω has 100 points between 0 to π .

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3. The difference equation between the output signal $y[n]$ and the input signal $x[n]$ is given by (10 marks)

$$y[n] = 2x[n] + 3x[n-1] - x[n-2] + x[n-3] + x[n-4] - x[n-5] + 3x[n-6] + 2x[n-7]$$

- a. Draw the filter structure in linear phase form.
- b. Find the phase response $\angle H(e^{j\omega})$.

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4. Butterworth lowpass filter has the following specification: (25 marks)

$$\Omega_p = 100, R_p = 3 \text{ dB}, \Omega_s = 250 \text{ and } A_s = 25 \text{ dB}$$

- Find N .
- Find Ω_c to obtain -3 dB at $\Omega_p = 100$ rads/sec.
- Find $H_a(s)$.
- Explain the concept to transform $H_a(s)$ to $H(z)$ by the impulse invariant method where T_s is the sampling period.
- Can we use the functions built in Matlab for problem d.? Give a reason.

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5. Digital lowpass filter has the following specification: (25 marks)

$$\omega_p = 0.2\pi, R_p = 1 \text{ dB}, \omega_s = 0.4\pi \text{ and } A_s = 20 \text{ dB}$$

Choose Chebyshev lowpass filter as the prototype filter and bilinear transformation method.

- Find Ω_p , Ω_s and N .
- Find the pole values.
- Find $H_a(s)$.
- Use functions in Matlab to find the frequency response $H(e^{j\omega})$ where ω has 100 points between 0 to π .
- How do we obtain the new $H(z)$ if ω_p is changed from 0.2π to be 0.3π ?

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For problem 5 continued

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6. The system function of an IIR filter is (5 marks)

$$H(z) = \frac{1 - 1.7321z^{-1} + z^{-2}}{1 - 1.3856z^{-1} + 0.64z^{-2}}$$

Draw the direct form II normal.

7. Answer the following questions. (10 marks)

- The signal given as $x[n] = \cos(0.5\pi n)$ where $0 \leq n \leq 63$ has the discrete Fourier Transform $|X(k)|$. What is the k value that $|X(k)|$ is maximum?
- The signal given as $x[n] = \cos(0.515625\pi n)$ where $0 \leq n \leq 63$ has discrete Fourier Transform $|X(k)|$. Is the maximum $|X(k)|$ at the frequency of the signal $x[n]$? Give a reason.