Digital Signal Processing

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Course Details

Course Code: 0113620

Course Name: Digital Signal Processing

(Sayısal İşaret İşleme)

Instructor : Nizamettin AYDIN

Assesment Method Quantity (%)Attendance& participation 05 Ouiz 05 3 Homework 15 Presentation 1 10 Midterm Exam(s) 1 25 Final Exam 40

Course Outline

1. Introduction.

Mathematical Representation of Signals. Mathematical Representation of Systems. 2. Sinusoids.

Review of Sine and Cosine Functions. Sinusoidal Signals. Sampling and Plotting Sinusoids. Complex Exponentials and Phasors. Phasor Addition. Time Signals.
 Spectrum Representation.
 The Spectrum of a Sum of Sinusoids. Beat Notes. Periodic Waveforms. Fourier Series Analysis and Synthesis. Time-Frequency Spectrum. Frequency Modulation.

Sampling and Aliasing.
 Sampling Spectrum View of Sampling and Reconstruction. Discrete-to-Continuous Conversion. The Sampling Theorem.

5. FIR Filters.

Discrete-Time Systems. The Running Average Filter. The General FIR Filter. Implementation of FIR Filters. Linear Time-Invariant (LTI) Systems. Convolution and LTI Systems. Cascaded LTI Systems. Example of FIR Filtering.

6. Frequency Response of FIR Filters.

Sinusoidal Response of FIR Systems. Superposition and the Frequency Response. Steady State and Transient Response. Properties of the Frequency Response. Graphical Representation of the Frequency Response. Cascaded LTI Systems. Running-Average Filtering. Filtering Sampled Continuous-Time Signals.

ransforms.

Definition of the z-Transform. The z-Transform and Linear Systems. Properties of the z-Transform. The z-Transform as an Operator. Convolution and the z-Transform. Practical Bandpass Filter Design. Properties of Linear Phase Filters.

Course Outline

8. IIR Filters.

The General IIR Difference Equation. Time-Domain Response. System Function of an IIR Filter. Poles and Zeros. Frequency Response of an IIR Filter. The Inverse z-Transform and Some Applications. Second-Order Filters. Frequency Response of Second-Order IIR Filter. Example of an IIR Lowpass Filter.

Continuous-Time Signals and LTI Systems.

Continuous-Time Signals and ETT Systems.

Continuous-Time Signals. The Unit Impulse. Continuous-Time Systems. Linear
Time-Invariant Systems. Impulse Responses of Basic LTI Systems. Convolution of
Impulses. Evaluating Convolution Integrals. Properties of LTI Systems.

10. The Frequency Response.

ne Frequency Response.

The Frequency Response Function for LTI Systems. Response to Real Sinusoi Signals. Ideal Filters. Application of Ideal Filters. Time-Domain or Frequency-Domain?

11. Continuous-Time Fourier Transform.

11. Continuous - Ime Fourier Transform.

Definition of the Fourier Transform. The Fourier Transform and the Spectrum. Examples of Fourier Transform Pairs. Properties of Fourier Transform Pairs. The Convolution Property. Basic LTI Systems. The Multiplication Property.

12. Filtering, Modulation, and Sampling.

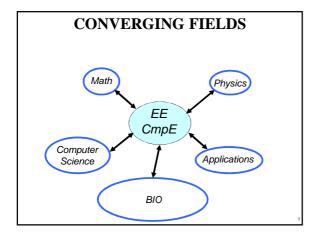
Linear Time-Invariant Systems. Sinewave Amplitude Modulation. Sampling and Reconstruction.

13. Computing the Spectrum.

Finite Fourier Sum. Time-windowing. Analysis of a Sum of Sinusoids. Dis Fourier Transform. Spectrum Analysis of Finite-Length Signals. Spectrum of Periodic Signals. The Spectrogram. The Fast Fourier Transform (FFT).

COURSE OBJECTIVE

- Students will be able to:
- Understand mathematical descriptions of signal processing algorithms
- · Express those algorithms as computer implementations (MATLAB or SCILAB)



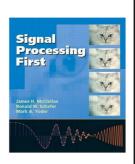
Main course book

Signal Processing First by James H McClellan, Ronald W. Schaffer

and Mark A. Yoder.

Published by Prentice Hall.

Isbn: 0-13-120265-0



some recommended books

- Understanding Digital Signal Processing by Richard G. Lyons
- The Scientist and Engineer's and Guide to Digital Signal Processing by Steven W. Smith.
- Digital Signal Processing and the Microcontroller by Dale Grover and John R. (Jack) Deller with illustrations by Jonathan Roth.
- Discrete-Time Signal Processing by A. V. Oppenheim and R. W. Schafer. Digital Signal Processing: Principles, Algorithms, and Applications by J. G. Proakis and D. G. Manolakis.
- Digital Signal Processing in Communication Systems by Marvin E. Frerking.

 Multirate Digital Signal Processing by R. E. Crochiere and L. R. Rabiner.
- Multirate Digital Signal Processing by R. E. Crochiere and L. R. Rabiner.
 Theory and Application of Digital Signal Processing by Rabiner and Gold. A comprehensive, industrial-strength DSP reference book.
 Digital Signal Processing by Alan V. Oppenheim and Ronald W. Schafer. Another industrial-strength reference.
 Digital Signal Processing by William D. Stanley.

Rules of the Conduct

- No eating /drinking in class
 - except water
- Cell phones must be kept outside of class or switched-off during class
 - If your cell-phone rings during class or you use it in any way, you will be asked to leave and counted as unexcused absent.
- · No web surfing and/or unrelated use of computers,
 - when computers are used in class or lab.

Rules of the Conduct

- You are responsible for checking the class web page often for announcements.
- · Academic dishonesty and cheating will not be tolerated and will be dealt with according to university rules and regulations
 - Presenting any work, or a portion thereof, that does not belong to you is considered academic dishonesty.
- University rules and regulations:
 - http://www.ogi.yildiz.edu.tr/category.php?id=17
 - https://www.yok.gov.tr/content/view/544/230/lang,tr_TR/

Attendance Policy

- The requirement for attendance is 70%.
 - Hospital reports are not accepted to fulfill the requirement for attendance.
 - The students, who fail to fulfill the attendance requirement, will be excluded from the final exams and the grade of F0 will be given.