

## ECEn 487. Introduction to Discrete-Time Signal Processing

<b>Catalog Description:</b>	<b>ECEn 487. Introduction to Discrete-Time Signal Processing.</b> (4:3:3) W Digital signal processing, fast Fourier transforms, digital filter design, spectrum analysis. Applications in speech processing, SONAR, communications, etc.	
<b>Course Type:</b>	Engineering Topics	
<b>Prerequisites:</b>	ECEn 370, 380	
<b>Textbooks and/or other required materials</b>	<i>Discrete Time Signal Processing, Second Edition</i> , A.V. Oppenheim and R.W. Schaffer with John R. Buck, Prentice Hall	
<b>Topics Covered:</b>	<ol style="list-style-type: none"> <li>1. Review of z-Transform and its properties.</li> <li>2. Inverse z-Transform using contour integration</li> <li>3. Sampling and multirate signal processing</li> <li>4. DFT, circular convolution, linear convolution from DFT</li> <li>5. The Fast Fourier Transform.</li> <li>6. Frequency response for rational system functions</li> <li>7. Relationship between magnitude and phase, all-pass systems, minimum phase, generalized linear phase.</li> <li>8. Filter design by impulse invariance and bilinear transform.</li> <li>9. Windowed design of FIR filters.</li> <li>10. Optimal approximations for FIR filter design.</li> <li>11. Discrete Hilbert Transforms.</li> <li>12. Hilbert Transforms by complex bandshift.</li> <li>13. Time-dependent Fourier Transform.</li> <li>14. Speech, SONAR, Adaptive Filtering, etc.</li> </ol>	
<b>Course Competencies:</b>	Application of integral calculus, discrete math, and complex variables, and transform theory to discrete-time signal processing.	Outcome 1
	Ability to apply probability theory in using histograms to estimate power spectral densities.	Outcome 1
	Ability to perform experiments with real-time DSP systems, including discrete-time filters, spectrum analyzer, adaptive filter, etc.	Outcome 2
	Completion of real-time DSP design projects, including discrete-time filters, spectrum analyzer, adaptive filter, etc.	Outcome 3
	Familiarity with basic DSP functions and algorithms (i.e. FIR filters, IIR filters, polyphase systems, windows, FFT's, etc.)	Outcome 5
	Ability to use basic DSP functions and algorithms (i.e., FIR filters, IIR filters, polyphase systems, windows, FFTs, etc.) to formulate and solve engineering problems.	Outcome 5
	Ability to use Matlab, C, and Code Composer Studio as tools to design, implement, and debug DSP functions.	Outcome 11
<b>Schedule:</b>	Lectures: One hour MWF Laboratory: Three hours per week TA Recitations: (None)	
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<b>Date:</b>	June 24, 2008	