

**ECE4634**  
**Digital Communications**  
**Fall 2007**

Instructor: Dr. R. Michael Buehrer  
Lecture #2: Review of Signals and  
the Fourier Transform



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**Overview**



- Information in communication systems is transferred through the use of EM waves
- At each point in the system, we observe signals. These signals can be described mathematically using both the time and the frequency domains.
- While the time domain is more familiar to most students, often the frequency domain is more intuitive for understanding certain signal characteristics
- At the receiver we observe both the desired waveform as well as undesired waveforms such as *noise* and *interference*.
- Reading
  - Sections 2.1-2.3

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**Objectives for this Lecture**



- Review important properties of signals
- Review a key mathematical tool for analyzing communication systems the *Fourier Transform*
  - Motivation for Fourier Theory
  - Common Fourier Transforms
  - Fourier Transform Properties

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## Course Objectives



- Design digital communication systems, given constraints on data rate, bandwidth, power, fidelity, and complexity;
- Analyze the performance of a digital communication link when additive noise is present in terms of the signal-to-noise ratio and bit error rate;
- Compute the power and bandwidth requirements of modern communication systems, including those employing ASK, PSK, FSK, and QAM modulation formats;
- Design a scalar quantizer for a given source with a required fidelity and determine the resulting data rate;
- Determine the auto-correlation function of a line code and determine its power spectral density;
- Determine the power spectral density of bandpass digital modulation formats.

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## Physically Realizable Waveforms



- Have finite time duration (finite energy!)
- Occupy finite frequency spectrum
- Are continuous
- Have finite peak value
- Are real-valued
- All real-world signals will have these properties, although sometimes we use mathematical models which violate these conditions.

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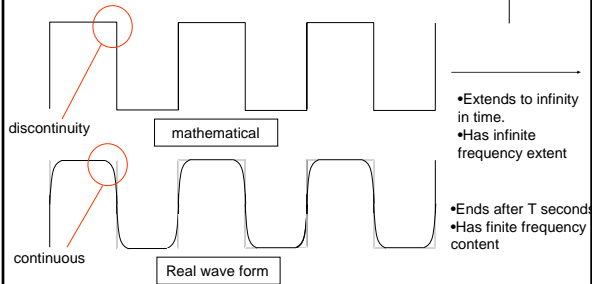
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## Mathematical Representations



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## Mathematical Representations



- Thus, when we analyze communication systems we use mathematical models.
- These models allow for convenient analysis but are not completely accurate concerning the real world.
- Fortunately, they provide close enough approximation that the conclusions reached using the models are still valid.

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## Energy and Power



- **Energy:**  $E = \int_{-\infty}^{\infty} w^2(t) dt$
- A signal  $w(t)$  is an Energy Signal if  $0 < E < \infty$
- **Power:**  $P = \langle w^2(t) \rangle = \lim_{T \rightarrow \infty} \frac{1}{T} \int_{-T/2}^{T/2} w^2(t) dt$
- For periodic signals, power can be computed by integrating over one period
- A signal  $w(t)$  is a Power Signal if  $0 < P < \infty$

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## Decibels



- Base 10 logarithmic measure of *power ratios*
- Useful when:
  - Power and energy levels vary over orders of magnitude
  - It is the ratio of two powers that is important
- When comparing power or energy:
$$dB = 10 \log_{10}(P_1/P_2)$$
- Sometimes it is useful to compare a power with 1 W or with 1 mW:

$$dBW = 10 \log_{10}(P_1/1W)$$
$$dBm = 10 \log_{10}(P_1/1mW)$$

For voltages or currents:

$$dB = 20 \log_{10}(V_1/V_2)$$
$$= 20 \log_{10}(I_1/I_2)$$

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## Complex Numbers



- A complex number is a number composed of two real numbers, one which represents the "real" part and one which represents the "imaginary" part (originally created for defining roots of a polynomial)

$$z = x + jy$$

Real part      Imaginary part

$\sqrt{-1}$

- We define addition and multiplication as

$$z = x + jy$$

$$c = a + jb$$

$$z + c = (x+a) + j(y+b)$$

$$z^*c = (xa - yb) + j(ya + xb)$$

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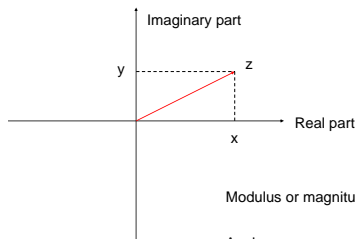
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## Complex Plane



Modulus or magnitude  $|z| = \sqrt{x^2 + y^2}$

Angle  $\arg(z) = \tan^{-1}\left(\frac{y}{x}\right)$

$$z = x + jy$$

$$= |z|(\cos \theta + j \sin \theta)$$

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## Why do we care?



- Answer:** Because of phase!
- In communications we often deal with sinusoidal signals
  - Further we may always represent a signal as the sum (or integral) of sinusoidal signals
  - Sinusoids are conveniently represented using complex numbers because of phase

$$x(t) = A \cos(\omega_c t + \theta)$$

$$= A \cos \theta \cos(\omega_c t) - A \sin \theta \sin(\omega_c t)$$

A convenient short hand is:  $\tilde{x}(t) = A e^{j\theta}$  where the frequency  $\omega_c$  is assumed

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## Using phasors



- The original signal is then represented as

$$x(t) = \text{Re} \{ \tilde{x}(t) e^{j\omega t} \}$$

- Note that while  $\tilde{x}(t)$  is complex, the true signal  $x(t)$  is always real. The complex nature is simply a convenient mathematical construct to readily handle phase components
- Further note that 
$$\tilde{x}(t) = A e^{j\theta} = A \cos \theta + j A \sin \theta$$
- This form will be particularly convenient when we analyze bandpass communication systems
- Note that while physical signals are always "real", the imaginary part of  $\tilde{x}(t)$  will have physical meaning as we will see later in the course

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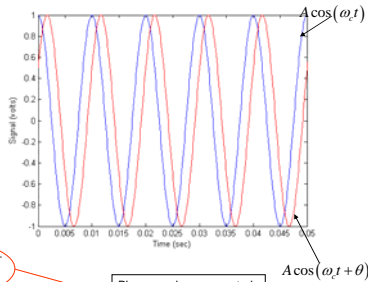
## Phase



- Phase represents time delay of a sinusoid

$$\begin{aligned} x(t) &= A \cos(\omega t) \\ x(t - t_0) &= A \cos(\omega(t - t_0)) \\ &= A \cos(\omega t - \omega t_0) \\ &= A \cos(\omega t - \theta) \end{aligned}$$

$$\begin{aligned} x(t) &= A \cos(\omega t) \\ x(t - t_0) &= A \cos(\omega t - \theta) \\ &= A \cos(\theta) \cos(\omega t) + A \sin(\theta) \sin(\omega t) \end{aligned}$$



Phase can be represented by a weighted sum of cos and sin

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## Euler's Identities



- $e^{jx} = \cos(x) + j \sin(x)$

- $\sin(x) = \frac{e^{jx} - e^{-jx}}{2j}$

- $\cos(x) = \frac{e^{jx} + e^{-jx}}{2}$

- Note:**  $|e^{jx}| = 1$

This is termed a *phasor* representation of sinusoidal signals. We will use this representation extensively.

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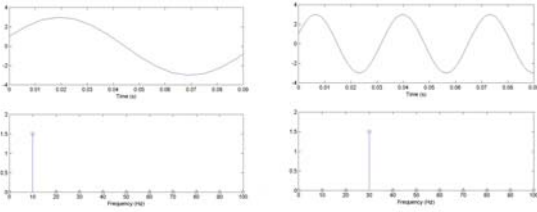
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## Frequency



- For a sinusoidal function the frequency is the inverse of the time it takes to complete one cycle (i.e., the period)



$f_0 = 10$

$$3\sin(2\pi f_0 t + \pi/9)$$

$f_0 = 30$

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## Frequency



- Any signal can be expressed as the weighted sum of sinusoids of varying frequencies and phases.
- Many physical objects respond to EM waves based on the frequency of the wave. Thus, we are interested in the frequency of signals.
- Fourier Theory allows us to view the 'frequency content' of a signal by decomposing the signal into an infinite sum (or an integral) of sinusoids.
- The Fourier Transform tells us how much of each frequency is needed.
  - Magnitude tells us the amount of each frequency
  - The phase tells us how much cos vs sin

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## The Fourier Transform



- One of the most common mathematical tools for analyzing a signal or waveform is the Fourier Transform.
- The Fourier Transform provides us with information concerning the *frequency content*.
- This is useful for:
  - Determining bandwidth
  - Demodulating frequency modulated signals
  - Understanding how objects or systems will respond to a signal (Transfer Function)
  - Equalization

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## The Fourier Transform



- The Fourier Transform of a signal  $w(t)$  is given by:

$$W(f) = F\{w(t)\} = \int_{-\infty}^{\infty} w(t)e^{-j2\pi ft} dt$$

- The Inverse Fourier Transform is given by:

$$w(t) = F^{-1}\{W(f)\} = \int_{-\infty}^{\infty} W(f)e^{j2\pi ft} df$$

- We denote a Fourier Transform pair by:  $w(t) \leftrightarrow W(f)$
- The Fourier Transform always exists if  $w(t)$  is an Energy Signal

We correlate the signal with a complex sinusoid of frequency  $f$  to determine how much of that frequency is present.

We sum complex sinusoids of different frequencies  $f$ , weighting them by the amount of each frequency contained by the signal.

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## Why do we use complex sinusoids?



- Answer: We are interested in *phase*
- If we defined the Fourier Transform as

$$W(f) = F\{w(t)\} = \int_{-\infty}^{\infty} w(t)\cos(2\pi ft) dt$$

we would lose any signal information related to  $\sin(\omega t)$

- If we only used  $\sin(\omega t)$  in the transform, we would lose any signal information related to  $\cos(\omega t)$
- Thus we must have both:  $e^{j\omega t} = \cos(\omega t) + j\sin(\omega t)$

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## Interpretation of Fourier Transform



- The Fourier Transform may be thought of as a tool for looking at a signal from a different perspective
  - Consider how different a chair might look when viewed from different angles



- Frequency measures the rate of change
  - High frequency corresponds to rapid change with time
  - Low frequency corresponds to slow change with time

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## Fourier Transform Example 2.1: Rectangular Pulse



$$\text{rect}\left(\frac{t}{T}\right) = \begin{cases} 1, & |t| \leq T/2 \\ 0, & |t| > T/2 \end{cases}$$

$$W(f) = \int_{-T/2}^{T/2} 1 \cdot e^{-j2\pi ft} dt = \frac{e^{-j\pi f T} - e^{j\pi f T}}{-j2\pi f}$$

$$\sin(x) = \frac{(e^{jx} - e^{-jx})}{2j}$$

$$W(f) = \frac{\sin(\pi f T)}{\pi f} = T \frac{\sin(\pi f T)}{\pi f T} = T \text{sinc}(f T)$$

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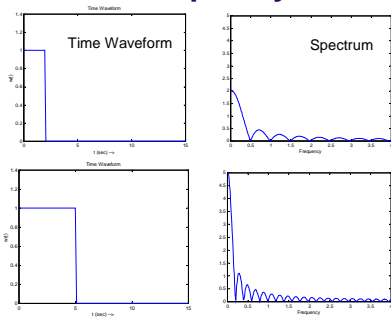
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## Time - Frequency



As we expand the duration of the pulse, the signal changes less rapidly. Thus the signal has more low frequency content.  
Time expands → Frequency compresses

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## Approach to Finding Fourier Transform Pairs



- We could continue to find transform pairs according to the definition, but this is inefficient
- In general, we compile a table of known transform pairs
- We also compile a table of simple rules for modifying transform pairs.
- Using the known pairs and transform properties we can find most transforms needed.

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## Fourier Transform Pairs



Rectangular Pulse	$\text{rect}\left(\frac{t}{T}\right)$	$T[\text{sinc}(fT)]$
Triangular Pulse	$\text{tri}\left(\frac{t}{T}\right)$	$T[\text{sinc}(fT)]^2$
Unit Step	$u(t)$	$\frac{1}{2}\delta(f) + \frac{1}{j2\pi f}$
Signum	$\text{sgn}(t)$	$\frac{1}{j\pi f}$
Constant	1	$\delta(f)$
Impulse at $t_0$	$\delta(t-t_0)$	$e^{-j2\pi ft_0}$
Sinc	$\text{sinc}(2Wt)$	$\frac{1}{2W}\text{rect}\left(\frac{f}{2W}\right)$
Phasor	$e^{j\omega_0 t + \phi}$	$e^{j\phi}\delta(f-f_0)$
Sinusoid	$\cos(2\pi ft + \phi)$	$\frac{1}{2}e^{j\phi}\delta(f-f_0) + \frac{1}{2}e^{-j\phi}\delta(f+f_0)$
Gaussian	$e^{-\pi(t/t_0)^2}$	$t_0 e^{-\pi(f/f_0)^2}$

Note: Think of a constant as a sinusoid with an infinite period ( $f = 0$ ). Does the transform make sense?

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## Important Transform Properties



- Linearity
- Time Delay
- Scale Change
- Duality
- Modulation
- Convolution
- Differentiation
- Integration

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## Fourier Transform Properties



Operation	Function	Fourier Transform
Linearity	$a_1 w_1(t) + a_2 w_2(t)$	$a_1 W_1(f) + a_2 W_2(f)$
Time Scale	$w(at)$	$\frac{1}{ a } W\left(\frac{f}{a}\right)$
Delay	$w(t-\tau)$	$W(f)e^{-j2\pi f\tau}$
Duality	$W(t)$	$w(-f)$
Real signal frequency translation	$w(t)\cos(2\pi ft + \theta)$	$\frac{1}{2}[e^{j\theta}W(f-f_0) + e^{-j\theta}W(f+f_0)]$
Complex Signal Frequency Translation	$w(t)e^{j2\pi f_0 t}$	$W(f-f_0)$
Bandpass Signal	$\text{Re}[g(t)e^{j2\pi f_0 t}]$	$\frac{1}{2}[G(f-f_0) + G^*(-f-f_0)]$
Differentiation	$\frac{d^n}{dt^n} w(t)$	$(j2\pi f)^n W(f)$
Integration	$\int_{-\infty}^t w(s)ds$	$(j2\pi f)^{-1} W(f) + \frac{1}{2}W(0)\delta(f)$
Convolution	$w_1(t) * w_2(t)$	$W_1(f)W_2(f)$
Multiplication	$w_1(t)w_2(t)$	$W_1(f) * W_2(f)$
Multiplication by $t$	$t^n w(t)$	$(-j2\pi f)^{-n} \frac{d^n W(f)}{df^n}$

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## In-class drill



- Find a mathematical expression for

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## Summary



- Today we have reviewed important concepts from signal and system theory that are particularly useful to communication system analysis
- Most importantly we will extensively use the Fourier Transform to analyze signals and systems in the *frequency domain*
- You should review
  - Fourier Transform pairs
  - Fourier Transform properties
- Next class we will examine useful functions termed *singularity functions*, review the concepts of Energy and Power Spectral Density, and the application of Fourier Theory to linear systems

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## Appendix

Additional Examples



Analogy and Digital Communications

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## Fourier Transform Example 2.2: Exponential



$$w(t) = u(t)e^{-at} = \begin{cases} e^{-at}, & t \geq 0 \\ 0, & t < 0 \end{cases}$$

$$\begin{aligned} W(f) &= \int_{-\infty}^{\infty} u(t)e^{-at}e^{-j2\pi ft} dt = \int_0^{\infty} e^{-(a+j2f\pi)t} dt \\ &= \left[ \frac{e^{-(a+j2f\pi)t}}{a+j2f\pi} \right]_0^{\infty} = \frac{1}{a+j2f\pi} \end{aligned}$$

- $|W(f)| = \frac{1}{\sqrt{a^2 + (2f\pi)^2}}$

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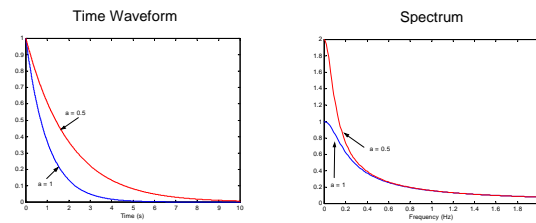
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## Example #2 (cont.)



As time waveform decreases more slowly, the more low frequency content in the wave.

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## Example 2.3: Cosine Waveform



- Easier to verify some transforms pairs from the inverse transform

$$w(f) = \frac{1}{2}\delta(f - f_c) + \frac{1}{2}\delta(f + f_c)$$

- $w(t) = \int_{-\infty}^{\infty} \frac{1}{2}\delta(f - f_c)e^{j2\pi ft} df + \int_{-\infty}^{\infty} \frac{1}{2}\delta(f + f_c)e^{j2\pi ft} df$   
 $= \frac{e^{j2\pi f_c t} + e^{-j2\pi f_c t}}{2}$   
 $= \cos(2\pi f_c t)$

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