

# Unit 1. Passband Modulation

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EL-GY 6013: DIGITAL COMMUNICATIONS

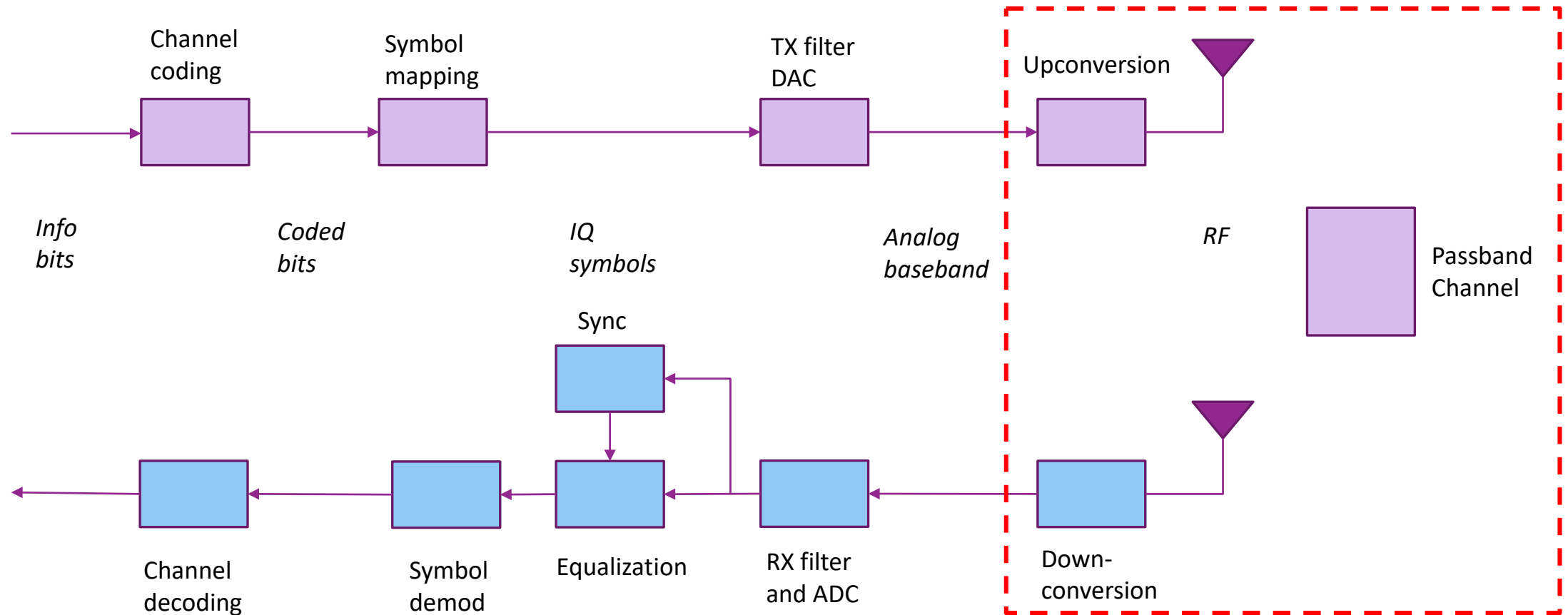
PROF. SUNDEEP RANGAN

# Learning Objectives

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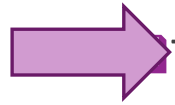
- ❑ Determine if a system is real passband or baseband
- ❑ Mathematically describe upconversion and downconversion
  - In time-domain and frequency-domain
- ❑ Compute simple continuous-time Fourier transforms (Review)
- ❑ Select parameters and analyze low-pass filter in down conversion
- ❑ Determine if a signal is a power or energy signal
  - Convert power in dBm
- ❑ Compute the effective baseband filter given a passband filter
- ❑ Model impairments such as time and frequency offsets

# This Unit



# Outline

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## Time-Domain Relationships

- ☐ Fourier Transform Review
- ☐ Frequency-Domain Relationships
- ☐ Power and Energy Spectra
- ☐ Baseband Equivalent Filters
- ☐ Wireless channels



# Signals in Communications

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□ **Signal:** Any quantity that varies in time

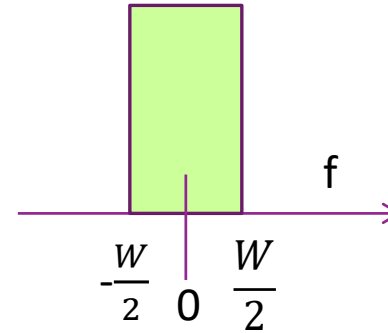
- Can be continuous time  $x(t)$
- Or discrete time  $x[n]$
- Real or complex valued

□ **Signals in communications:**

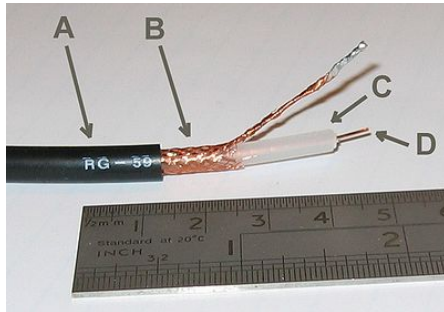
- $v(t)$  = Voltage at a particular point / place in a circuit (relative to ground)
- $E_z(t)$  = Electric field strength in a particular direction  
Note: electric field is a vector quantity  $E(t) = [E_x(t), E_y(t), E_z(t)]$
- A digital sample of a signal
- An intermediate value used in processing a signal

# Real Baseband Systems

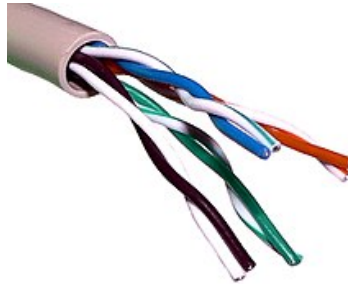
- Real baseband communication systems:
  - Communicate with lowpass real-valued signals
  - $X(f) \approx 0$  for  $|f| \leq \frac{W}{2}$



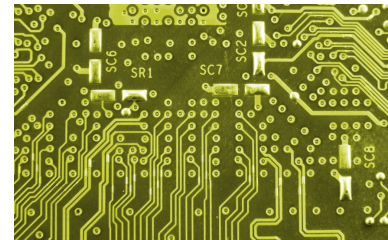
## Examples



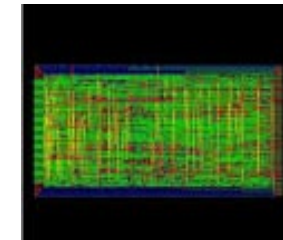
Coaxial cable



Twisted pair



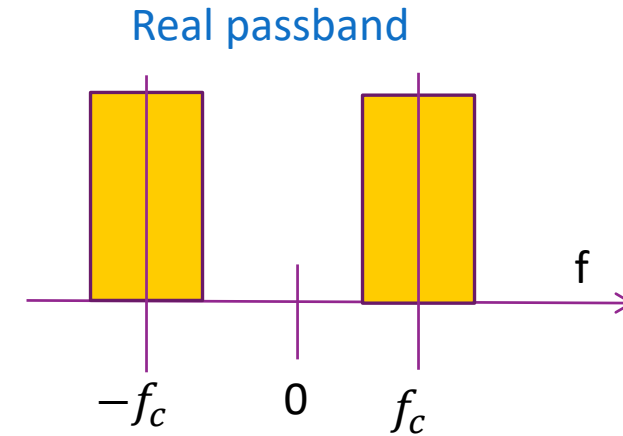
PCB traces  
e.g. microstrip  
or stripline



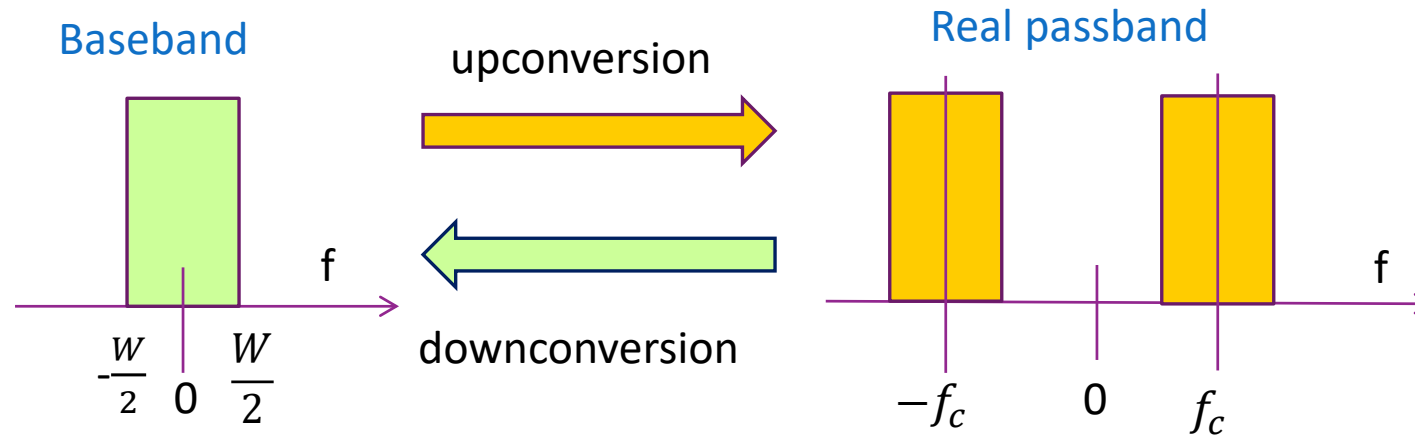
ASIC metal  
traces

# Real Passband Communications

- ❑ Real passband communication systems
  - Transmit around a **carrier frequency**  $f_c$
  - $f_c$  is sometimes called the **center frequency**
  - $X(f) \neq 0$  for  $|f - f_c| < W$  and  $|f + f_c| < W$
- ❑ Mostly radio frequency communication
  - Often wireless
  - Transmissions are restricted to bandwidth
  - Also, RF propagation is limited to certain bands
  - RF communication also occurs over cables



# Up- and Downconversion



- ❑ Up and downconversion: Shift center frequency of signals
- ❑ Used for all passband communications systems
  - Information occurs or is processed in **baseband**
  - Transmitted and received in **real passband**



# Upconversion in Time Domain

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## ❑ Baseband signals: $u_i(t)$ and $u_q(t)$ ,

- Also called “in-phase” and “quadrature” (I and Q)
- Real-valued. Typically bandlimited to  $|f| < \frac{W}{2}$  ( $\frac{W}{2}$  = Single-sided bandwidth)
- Sometimes called the “cosine” and “sine” part.

## ❑ Carrier frequency $f_c$

- Also called the “center” frequency

## ❑ Create real passband signal:

$$u_p(t) = u_i(t) \cos(2\pi f_c t) - u_q(t) \sin(2\pi f_c t)$$

## ❑ Upconversion is also called modulation

- But, we will use that term for something later.

# Downconversion

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❑ Can recover I part from multiplication by sinusoid:

❑ Recovery of the I part:

- $v_i(t) = 2u_p(t) \cos(2\pi f_c t) = u_i(t) + \text{high freq terms}$
- $u_i(t) = \text{LPF}(v_i(t))$

❑ Recovery of the Q part:

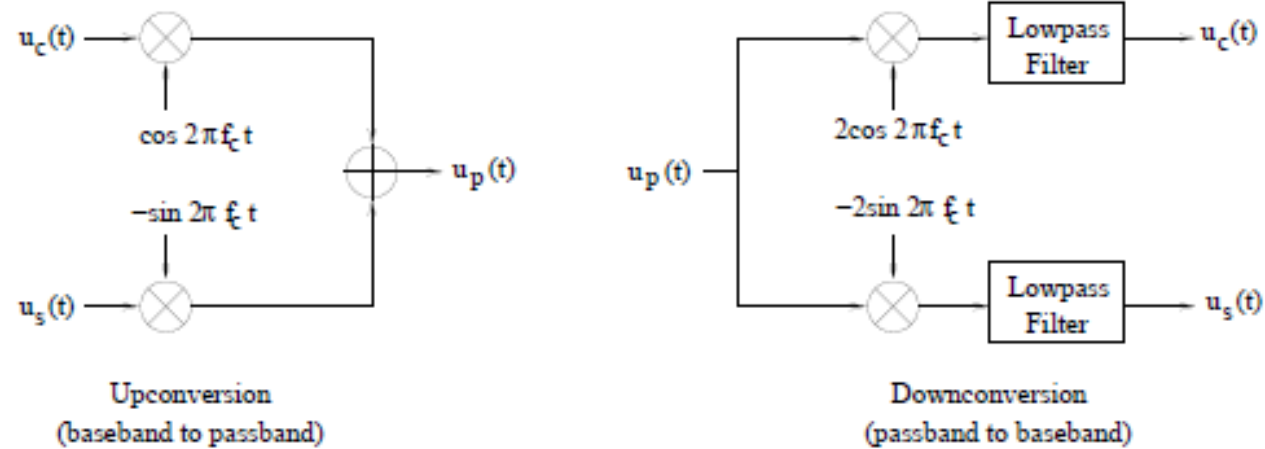
- $v_q(t) = -2u_p(t) \sin(2\pi f_c t) = u_q(t) + \text{high freq terms}$
- $u_q(t) = \text{LPF}(v_q(t))$

❑ Can derive relations using

$$\sin(2x) = 2 \sin(x) \cos(x) \quad 2\cos^2(x) = 1 + \cos(2x)$$

❑ Note gain of 2 and sign.

# Up and Downconversion Block Diagram



- Fig. 2.28 from Madhow
- Implementation with multipliers

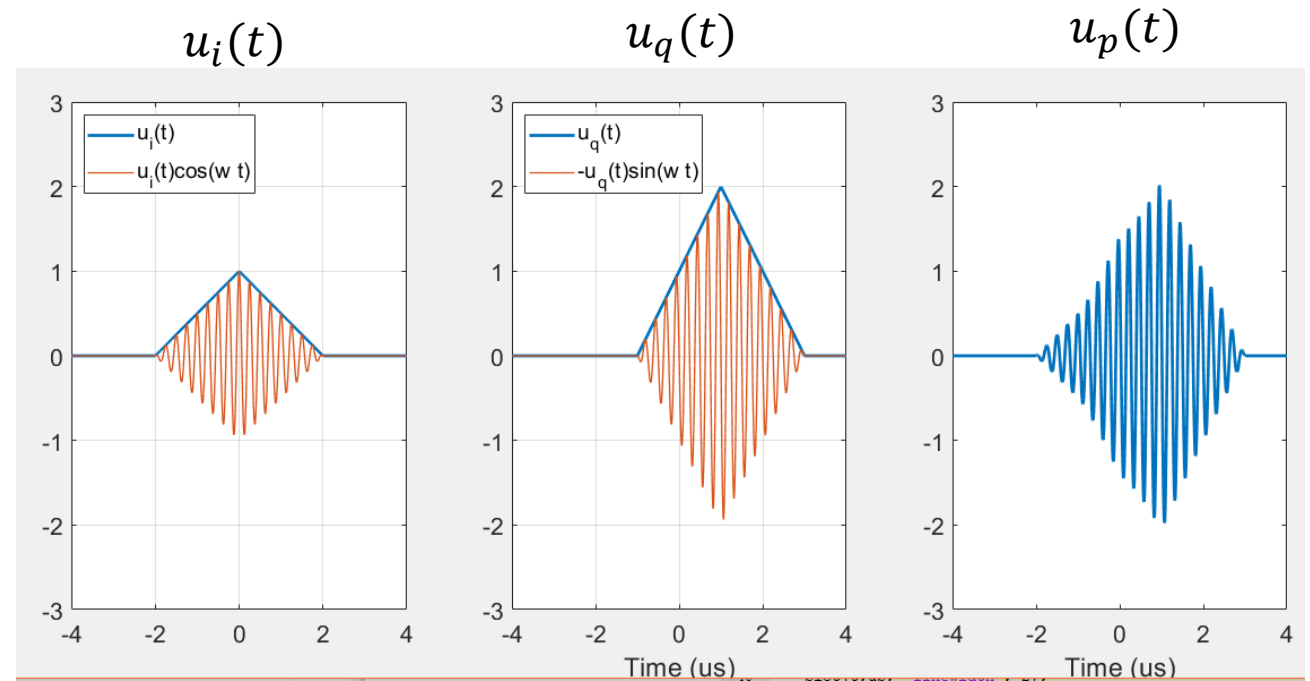
# Example

□ Suppose

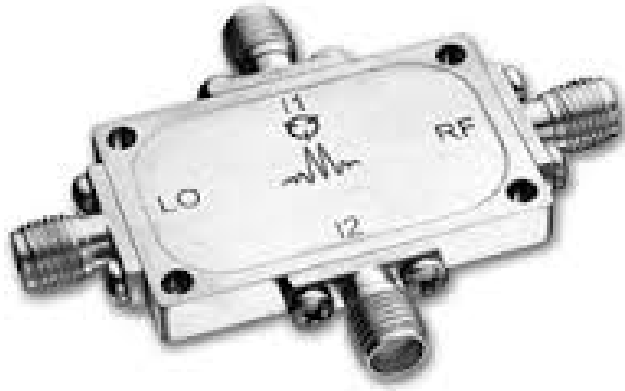
- $u_i(t) = \text{Tri}\left(\frac{t}{T}\right)$ ,  $u_q(t) = 2\text{Tri}\left(\frac{t}{T} - 0.5\right)$ ,  $\text{Tri}(s) := \max(0, 1 - |s|)$
- $T = 2 \mu\text{s}$ ,  $f_c = \frac{8}{T} = 4 \text{ MHz}$

```
% Create the baseband signals
nt = 1024;
T = 2.0;
t = linspace(-2*T, 2*T, nt)';
f0 = 8/T;
ui = max(1-abs(t/T), 0);
uq = 2*max(1-abs(t/T-0.5), 0);

% Modulate the I and Q components
uicos = ui.*cos(2*pi*f0*t);
uqsin = -uq.*sin(2*pi*f0*t);
up = uicos + uqsin;
```



# Actual IQ Mixer



- ❑ LO = “local oscillator” = square or sine wave at  $f_c$
- ❑ I1, I2 = I and Q inputs.
  - Generally, lowpass
- ❑ RF = passband output centered at  $f_c$

[http://www.markimicrowave.com/Mixers/IQ\\_Quadrature-IF\\_Double-Balanced/IQ-0318.aspx](http://www.markimicrowave.com/Mixers/IQ_Quadrature-IF_Double-Balanced/IQ-0318.aspx)

Datasheet	RF [GHz]	LO [GHz]	IF [MHz]	Conversion Loss [dB]	Image Rejection [dB]	Amplitude Deviation [dB]	Phase Deviation [Degrees]	Isolation L-R [dB]	Isolation L-I [dB]
<a href="#"><u><b>IQ-0318</b></u></a>	3 to 18	3 to 18	DC to 500	7	22	0.75	10	40	20

# Complex Envelope

□ Complex envelope:

$$u(t) = u_i(t) + ju_q(t)$$

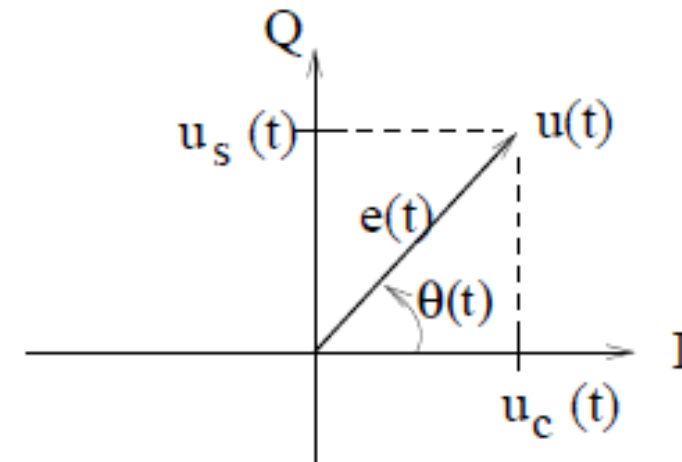
□ Magnitude and phase:

- $e(t) = \sqrt{u_c^2(t) + u_s^2(t)}$
- $\theta(t) = \tan^{-1}(u_s(t)/u_c(t))$

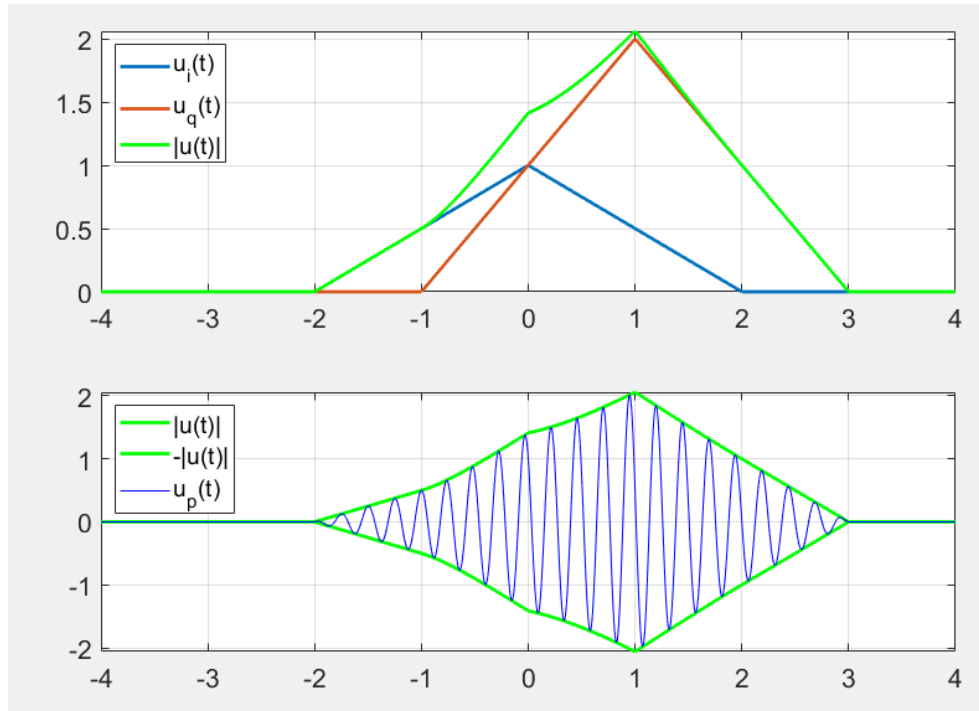
□ Can recover passband signal:

$$u_p(t) = e(t)\cos(2\pi f_c t + \theta(t))$$

Fig 2.29, Madhow



# Visualizing the Convex Envelope



□ Observe passband signal:

- Oscillates between  $-|u(t)|$  and  $|u(t)|$
- Is much higher frequency than baseband

```
%% Plot the convex envelope
umag = abs(ui + li*uq);

subplot(2,1,1);
plot(t,[ui uq],'Linewidth', 2);
hold on;
plot(t,umag,'g-','Linewidth',2);
hold off;
grid on;
legend('u_i(t)', 'u_q(t)', '|u(t)|','Location','NorthWest');
set(gca,'FontSize',16);

subplot(2,1,2);
plot(t,[umag -umag],'g-','Linewidth',2);
hold on;
plot(t,up,'b-');
hold off;
legend('|u(t)|', '-|u(t)|', 'u_p(t)','Location','NorthWest');
grid on;
set(gca,'FontSize',16);
```

# Complex Baseband has all Information

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- ❑ Can recover passband signal from alternate form:

$$u_p(t) = \text{Re}[u(t)e^{2\pi j f_c t}]$$

- Derive on board

- ❑ All information content of the signal is in  $u(t)$

- The passband signal is the complex baseband with a fast, but predictable offset



# Complex Notation for Downconversion

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□ Upconversion:  $u_p(t) = \text{Re}(u(t)e^{j\omega_c t})$

□ Downconversion:

- $v(t) = 2u_p(t)e^{-j\omega_c t} = u(t) + \text{High freq terms}$
- $u(t) = H_{LPF}(v(t))$

□ Proof:

- $u_p(t) = \text{Re}(u(t)e^{j\omega_c t}) = \frac{1}{2}(ue^{j\omega_c t} + u^*e^{-j\omega_c t})$
- $2u_p(t)e^{-j\omega_c t} = u(t) + \text{HFT}$
- HFT = high frequency terms

# Sample Problem (Soln on Board)

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□ Suppose that  $T = 1 \mu s$  and

$$u(t) = \begin{cases} 1 + j & t \in [0, T) \\ 1 - j & t \in [T, 2T) \\ 0 & \text{else} \end{cases}$$

□ What are  $u_i(t)$  and  $u_q(t)$ ? Draw them.

□ Write an equation for  $u_p(t)$  with a carrier frequency  $f_c = 4 \text{ MHz}$

□ Draw  $u_p(t)$  for  $t \in [1, 2] \mu s$ .

# Outline

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☐ Time-Domain Relationships

☐ Fourier Transform Review

 ☐ Frequency-Domain Relationships

☐ Power and Energy Spectra

☐ Baseband Equivalent Filters

☐ Wireless channels

# Fourier Transform

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□  $s(t)$ : real or complex continuous-time signal

□ Fourier Transform: time-domain to frequency domain

$$S(f) = \int_{-\infty}^{\infty} s(t) e^{-2\pi i f t} dt$$

□ Inverse Fourier transform:

$$s(t) = \int_{-\infty}^{\infty} S(f) e^{2\pi i f t} df$$

□ Represents signals in their frequency components

# FT in Angular Frequency

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□ Angular frequency:  $\omega = 2\pi f$

□ Fourier Transform: time-domain to frequency domain

$$S(\omega) = \int_{-\infty}^{\infty} s(t)e^{-i\omega t} dt$$

□ Inverse Fourier transform:

$$s(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} S(\omega)e^{i\omega t} d\omega$$

□ Note scaling

□ Some texts use other scalings

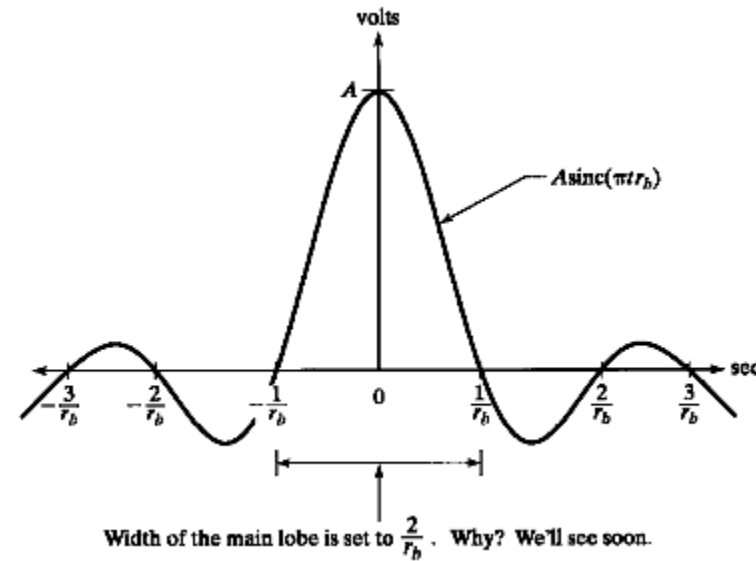
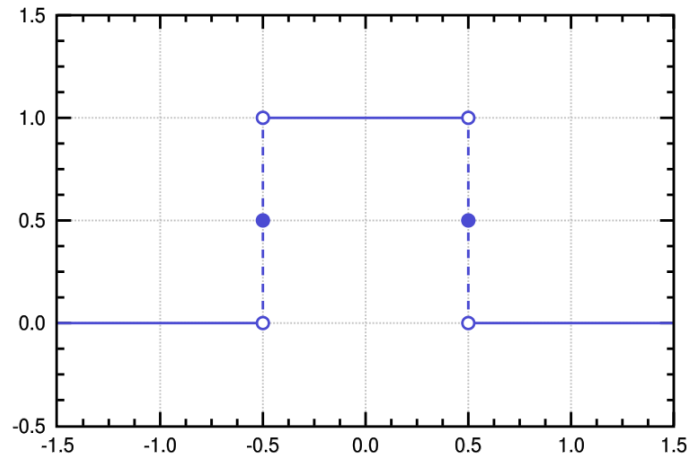
# Rect and Sinc

$$\square \text{rect}(at) \leftrightarrow \frac{1}{|a|} \text{sinc}\left(\frac{f}{a}\right) = \frac{\sin(\pi f/|a|)}{\pi f}$$

$$\square \text{sinc}(at) \leftrightarrow \frac{1}{|a|} \text{rect}\left(\frac{f}{a}\right)$$

$$\text{Height} = \frac{1}{|a|}$$

$$\text{Main lobe } f = \pm|a|$$



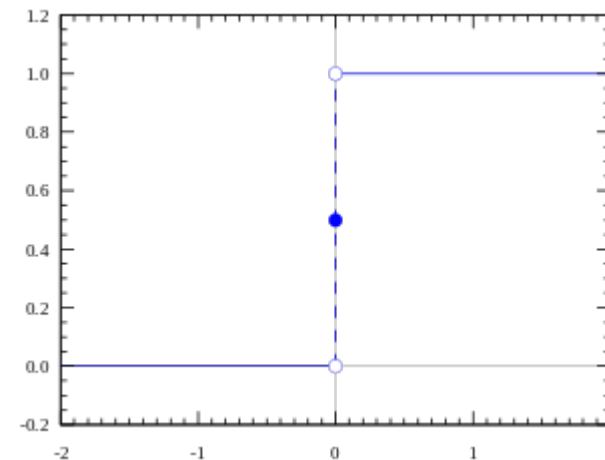
# Unit Steps

## Unit step

$$u(t) = \begin{cases} 1 & t > 0 \\ 1/2 & t = 0 \\ 0 & t < 0 \end{cases}$$

## FT:

- $e^{-\alpha t}u(t) \leftrightarrow \frac{1}{\alpha + 2\pi i f}, \operatorname{Re}(\alpha) > 0$
- $u(t) \leftrightarrow \frac{1}{2}\delta(f) + \frac{1}{2\pi i f}$



# Table (From WikiPedia)

	Function	Fourier transform unitary, ordinary frequency
	$f(x)$	$\hat{f}(\xi) = \int_{-\infty}^{\infty} f(x) e^{-2\pi i x \xi} dx$
201	$\text{rect}(ax)$	$\frac{1}{ a } \cdot \text{sinc}\left(\frac{\xi}{a}\right)$
202	$\text{sinc}(ax)$	$\frac{1}{ a } \cdot \text{rect}\left(\frac{\xi}{a}\right)$
203	$\text{sinc}^2(ax)$	$\frac{1}{ a } \cdot \text{tri}\left(\frac{\xi}{a}\right)$
204	$\text{tri}(ax)$	$\frac{1}{ a } \cdot \text{sinc}^2\left(\frac{\xi}{a}\right)$
205	$e^{-ax} u(x)$	$\frac{1}{a + 2\pi i \xi}$
206	$e^{-\alpha x^2}$	$\sqrt{\frac{\pi}{\alpha}} \cdot e^{-\frac{(\pi \xi)^2}{\alpha}}$
207	$e^{-a x }$	$\frac{2a}{a^2 + 4\pi^2 \xi^2}$

$f(x)$	$\hat{f}(\xi) = \int_{-\infty}^{\infty} f(x) e^{-2\pi i x \xi} dx$
1	$\delta(\xi)$
$\delta(x)$	1
$e^{iax}$	$\delta\left(\xi - \frac{a}{2\pi}\right)$
$\cos(ax)$	$\frac{\delta\left(\xi - \frac{a}{2\pi}\right) + \delta\left(\xi + \frac{a}{2\pi}\right)}{2}$
$\sin(ax)$	$\frac{\delta\left(\xi - \frac{a}{2\pi}\right) - \delta\left(\xi + \frac{a}{2\pi}\right)}{2i}$

$\text{sgn}(x)$	$\frac{1}{i\pi\xi}$
$u(x)$	$\frac{1}{2} \left( \frac{1}{i\pi\xi} + \delta(\xi) \right)$
$\sum_{n=-\infty}^{\infty} \delta(x - nT)$	$\frac{1}{T} \sum_{k=-\infty}^{\infty} \delta\left(\xi - \frac{k}{T}\right)$





# Other Properties

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$$\square s(t - a) \leftrightarrow e^{-2\pi i a f} S(f)$$

$$\square e^{2\pi i a t} s(t) \leftrightarrow S(f - a)$$

$$\square s(at) \leftrightarrow S(f/a)/|a|$$

$$\square d^n s(t)/dt^n \leftrightarrow (2\pi i f)^n S(f)$$

$$\square t^n s(t) \leftrightarrow d^n S(f)/df^n$$

$$\square s^*(t) \leftrightarrow S^*(-f)$$

$$\square s(t) \leftrightarrow S(f) \Rightarrow S(t) \leftrightarrow s(-f)$$

# Problems

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2.10 Determine the Fourier transform of each of the following signals ( $\alpha$  is positive).

1.  $x(t) = \frac{1}{1+t^2}$

2.  $\Pi(t-3) + \Pi(t+3)$

3.  $\Lambda(2t+3) + \Lambda(3t-2)$

4.  $\text{sinc}^3 t$

5.  $t \text{sinc } t$

6.  $t \cos 2\pi f_0 t$

7.  $e^{-\alpha|t|} \cos(\beta t)$

8.  $te^{-\alpha t} \cos(\beta t)$

□ Solutions on board

# Outline

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☐ Time-Domain Relationships

☐ Fourier Transform Review

 ☐ Frequency-Domain Relationships

☐ Power and Energy Spectra

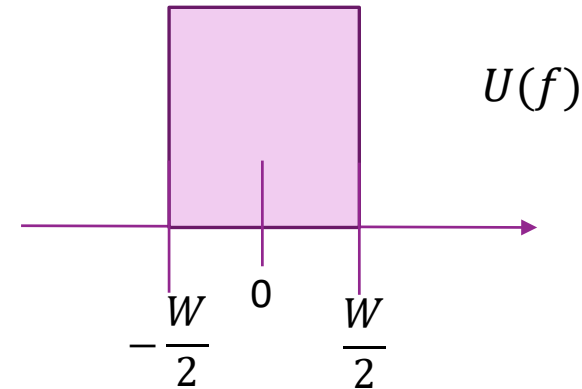
☐ Baseband equivalent filters

☐ Wireless channels

# Bandwidth Terminology

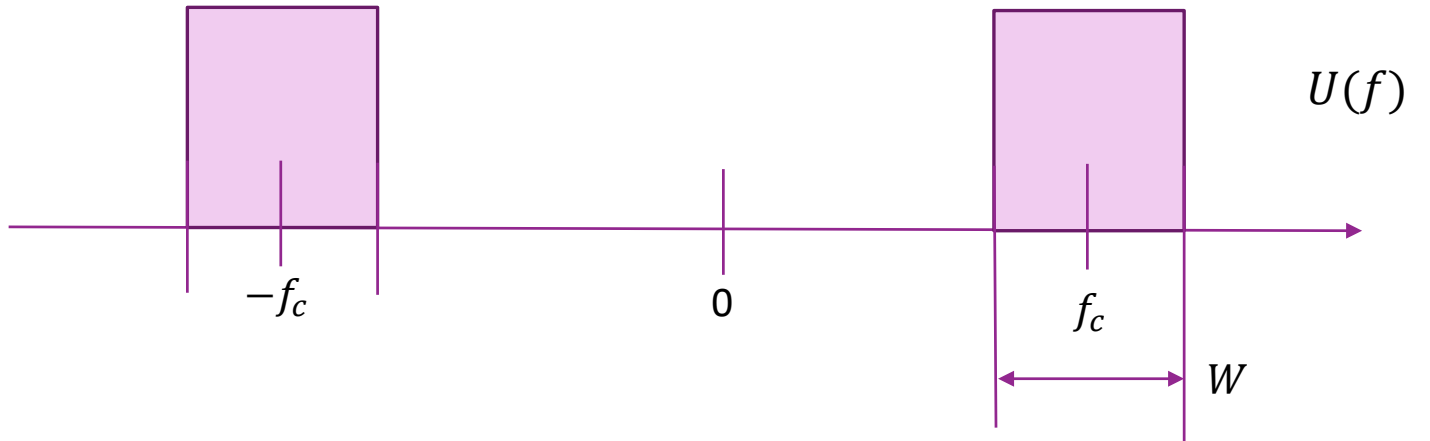
## □ Baseband signals

- Centered around  $f = 0$ , complex
- $\frac{W}{2}$  = single sided bandwidth
- $W$  = two sided bandwidth
- Band-limited to  $|f| \leq \frac{W}{2}$



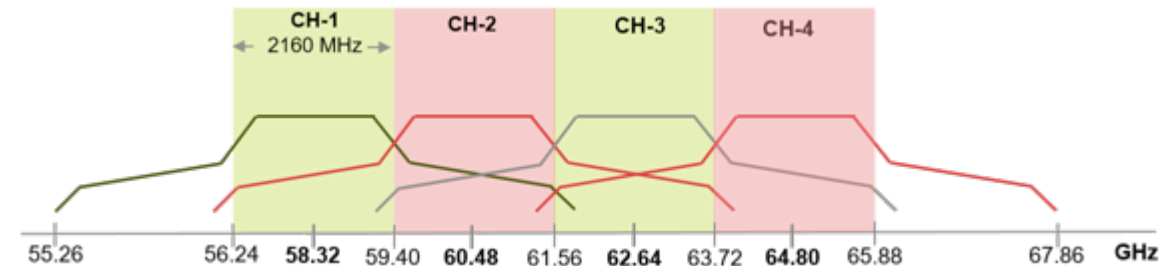
## □ Passband signals

- Centered around  $f = f_c$ , real
- $W$  = bandwidth (per side or image)
- Band-limited to  $|f - f_c| \leq \frac{W}{2}$



# Importance of Bandwidth

- ❑ Data rate generally scales linearly in bandwidth
  - If the transmit power and bandwidth increase by  $N \Rightarrow$  the communication rate increase by  $N$
  - We will see this in detail later
- ❑ Ex: Compare GSM (2G) and LTE (4G)
  - Single channel of GSM system = 200 kHz
  - Single channel of LTE = 20 MHz
  - If power scales sufficiently, LTE would in general have 100x data rate
  - LTE, in fact, can have even more capacity due to other improvements
- ❑ Figure to the right: 802.11ad channels
  - The channels are > 2 GHz



# Frequency Domain Relationships

## Baseband to Passband

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□ Suppose that  $U(f)$  is bandlimited to  $[-W, W]$  and  $f_c > W$

$$U_p(f) = \frac{1}{2} [U(f - f_c) + U^*(-f - f_c)]$$

□ Use notation:

- $U^+(f) := \frac{1}{2} U(f - f_c)$ : This is  $U(f)$  shifted to the right by  $f_c$  and scaled by  $\frac{1}{2}$
- $U^-(f) := \frac{1}{2} U^*(-f - f_c)$ : Flip  $U^+(f)$  around  $y$  axis and take negative of the imaginary part

□ Proof:

- Let  $c(t) = u(t)e^{2\pi j f_c t} \leftrightarrow C(f) = U(f - f_c)$
- $u_p(t) = \text{Re}(c(t)) = \frac{1}{2}(c(t) + c^*(t))$
- Now use conjugate symmetry  $c^*(t) \leftrightarrow C^*(-f)$

# Example Problem

□ Suppose baseband signal is as drawn:

□ What is the:

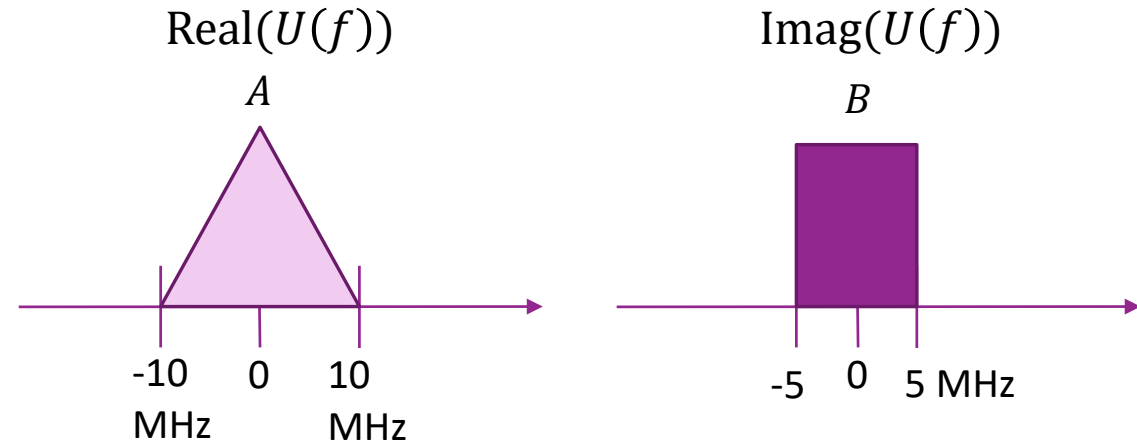
- Single-sided bandwidth?
- Two-sided bandwidth?

□ Write an equation for  $u_i(t)$

- You do not need to evaluate the integral.

□ Draw the passband frequency response if  $f_c = 2$  GHz

- Draw both the positive and negative images



# Frequency Domain Relationships

## Passband to Baseband

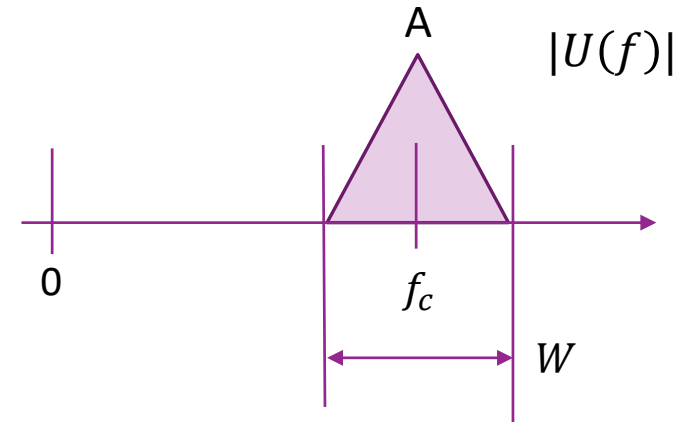
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- ❑ Downconversion in time-domain:  $v(t) = 2u_p(t)e^{-j\omega_c t}$ ,  $u(t) = h_{LPF}(t) * v(t)$
- ❑ In frequency-domain:  $U(f) = 2U_p(f + f_c)H_{LPF}(f)$ 
  - Shift to left, scale by 2 and filter
- ❑ Ideal filtering:
  - Suppose  $U_p(f)$  has bandwidth  $W$  around  $f_c$
  - Then typically have:  $H_{LPF}(f) = 1$  for  $|f| \leq \frac{W}{2}$  and  $H_{LPF}(f) = 0$  for  $|f| > \frac{W}{2}$
  - $U(f) = 2U_p(f + f_c)1_{\{|f| \leq W\}}$
  - Shift to the left and remove left image.
- ❑ Pictures on board

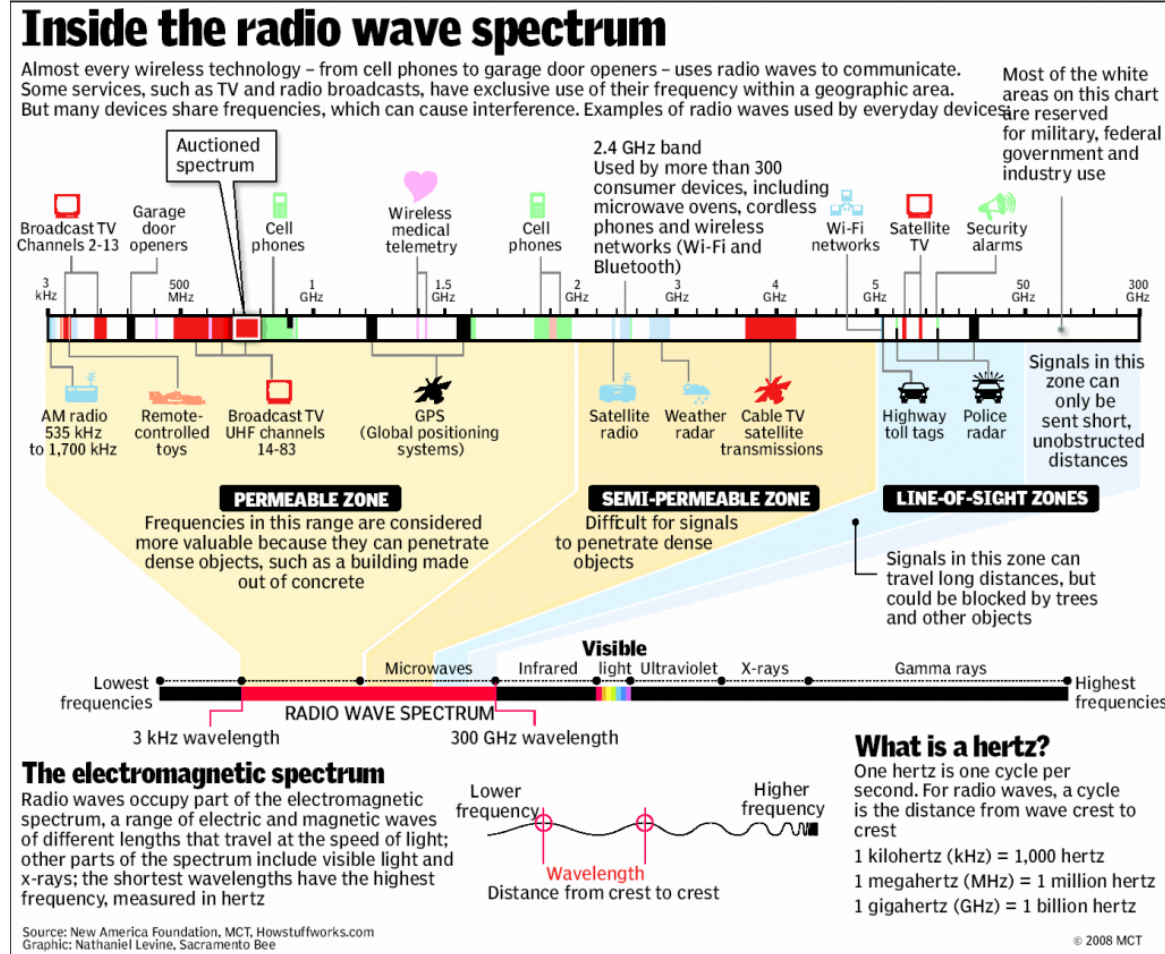


# Example Problem

- Suppose right image of passband is as shown:
  - $W = 4$  MHz,  $f_c = 800$  MHz
- Assume a LPF  $H_{LPF}(f) = \text{Rect}\left(\frac{f}{f_0}\right)$
- Draw magnitude spectrum of down-converted signal
  - When  $f_0 = 5$  MHz
  - When  $f_0 = 3$  MHz
- What range of values  $f_0$  will:
  - Keep the low-pass component
  - Reject the high frequency component
- Solution on board



# Radio Spectrum



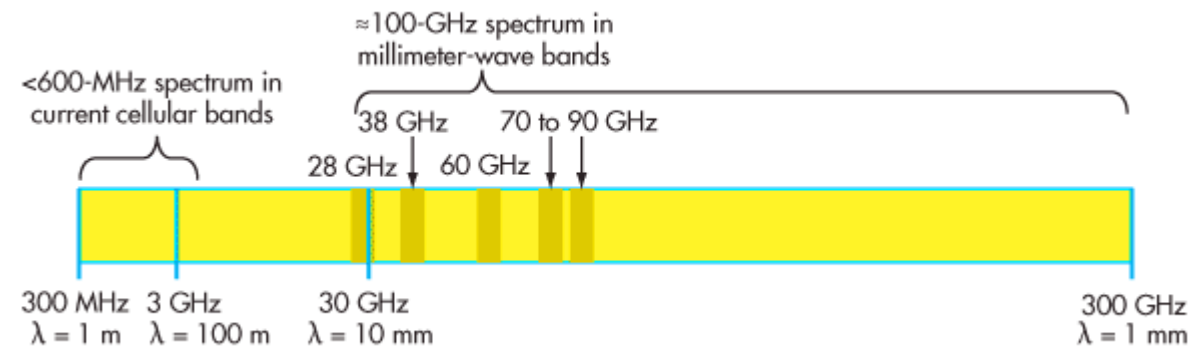
# Bandwidth and Center Frequencies Examples

System	Duplex	Center freq (MHz)	Bandwidth
GSM	FDD	GSM-850: 824-849 (UL), 869-894 (DL) GSM-900: 890-914 (UL), 935-959 (DL) GSM-1800: 1710–1784(UL), 1805.2–1879(DL) GSM-1900: 1850–1910(UL), 1930–1990(DL)	200 kHz per channel
UMTS	FDD	GSM + other bands ~2100 and ~1900	5 MHz per carrier
LTE	Mostly FDD	Mostly in 2100 to 2600 MHz	1.4 to 20 MHz, 10 MHz typical
802.11abg	TDD	2.4 GHz (ISM band) and 5 GHz (U-NII band)	20 MHz
802.11n			20, 40 MHz
802.11ac			20-160 MHz
802.11ad	TDD	60 GHz (millimeter wave spectrum)	2.16 GHz



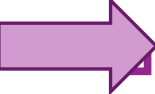
# Millimeter Wave

- ❑ New bands for 5G
  - 100x more bandwidth than conventional bands below 6 GHz
  - Bands at 28 GHz and 38 GHz opened up by FCC
  - 5G systems operating in these bands are coming soon



# Outline

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- ☐ Time-Domain Relationships
- ☐ Fourier Transform Review
- ☐ Frequency-Domain Relationships
-  ☐ Power and Energy Spectra
- ☐ Baseband equivalent filters
- ☐ Wireless channels

# Energy and Power Signals

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□ Instantaneous power:  $|x(t)|^2$

- Why squared?

□ Energy:

- $E_x = \int_{-\infty}^{\infty} |x(t)|^2 dt$
- Signal is called an “energy signal” if  $E_x < \infty$

□ Power:

- $P_x = \lim_{T \rightarrow \infty} \frac{1}{T} \int_{-T/2}^{T/2} |x(t)|^2 dt$
- Energy per unit time
- Signal is called a “power signal” if limit  $P_x$  exists and is finite

# Power of a Periodic Signal

## Time-Domain Method

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□ Suppose  $x(t)$  is periodic, period  $T$

□ **Theorem:**  $x(t)$  is a power signal and power can be computed from any one period

$$P_x = \frac{1}{T} \int_0^T |x(t)|^2 dt = \frac{1}{T} \int_{t_0}^{t_0+T} |x(t)|^2 dt$$

- Proof on board

# Example: Done on board

---

□ Suppose that  $x(t)$  has period  $T$

$$x(t) = a + bt, t \in [0, T]$$

- $a, b$  are real

□ Draw  $x(t)$

□ What is  $P_x$

□ What if  $a, b$  were complex?



# Power of a Periodic Signal

## Fourier Series Method

---

□ Suppose that  $x(t)$  is periodic with period  $T$

□ Then has Fourier Series

$$x(t) = \sum c_n e^{2\pi j f_n t}, \quad f_n = n/T$$

□ Theorem: Power of  $x(t)$  is:

$$P_x = \sum |c_n|^2$$

□ Note that if  $x(t) = \sum g(t - nT)$ , then  $c_n = G(f_n)$

- Can compute power from Fourier transform of  $g(t)$

# Example: On board

---

□ Suppose  $T = 10 \mu s$ ,

$$x(t) = \sum g(t - nT), \quad g(t) = \begin{cases} 2 & t \in [0, T/4) \\ -1 & t \in [T/4, T) \\ 0 & \text{else} \end{cases}$$

□ Draw  $x(t)$

□ What is the FT  $G(f)$ ?

□ What is the FT  $X(f)$ ?

□ What is the power of  $x(t)$ ?

□ What fraction of power of  $x(t)$  is in the  $|f| \leq 250 \text{ kHz}$ ?

# Energy Density

---

□ Energy of signal:  $E_x = \int |x(t)|^2 dt$

□ From Parseval's identity:  $E_x = \int |X(f)|^2 df$

- Can compute energy in frequency-domain

□ Energy density:  $G_x(f) = |X(f)|^2$

- Density of energy around frequency  $f$

# Power Spectral Density (PSD)

---

- ❑ Three equivalent ways to define PSD
- ❑ Definition 1: via windowing in time
- ❑ Definition 2: via filtering
- ❑ Definition 3: via auto-correlation for a random process
  - More advanced.
  - We will cover this in the next unit

# PSD: Time-Windowing Definition

---

□ Let  $x(t)$  be a power signal

□ Define windowed signal:

$$x_T(t) = \begin{cases} x(t) & |t| \leq T \\ 0 & |t| > T \end{cases}$$

□ PSD is defined as:

$$S_x(f) := \lim_{T \rightarrow \infty} \frac{1}{2T} |X_T(f)|^2$$

□ Similar to energy signal, but with averaging over time.

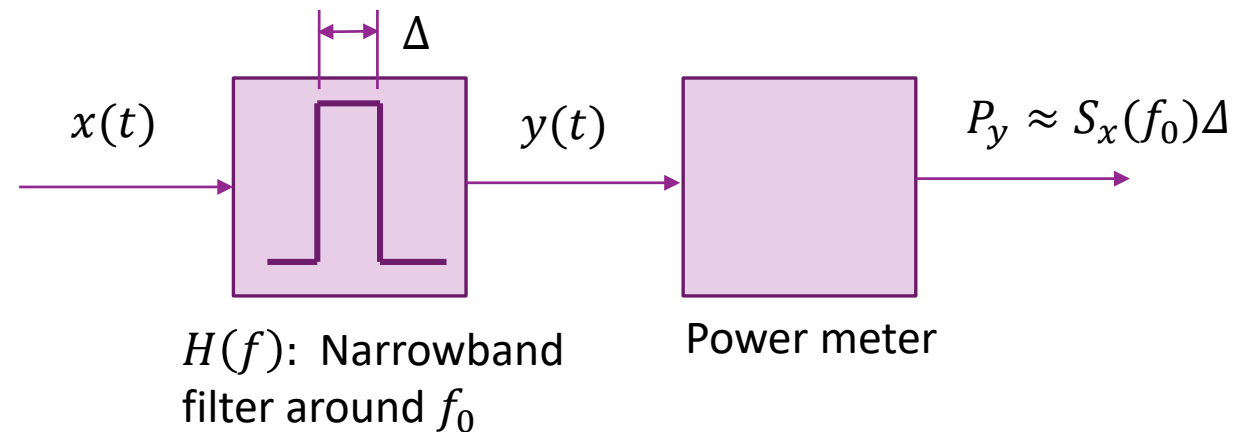
□ Can show power is given by:

$$P_x = \int_{-\infty}^{\infty} S_x(f) df$$

- $S_x(f)$  represents power per unit frequency

# PSD: Filtering Definition

- ❑ Let  $x(t)$  be a power signal
- ❑ Select frequency  $f_0$  to measure PSD
- ❑ Filter with narrowband filter
  - $y(t) = h(t) * x(t)$
  - $H(f) = 1$  for  $|f - f_0| \leq \Delta/2$
- ❑ Measure power  $P_y$



- ❑ PSD at  $f_0$  is defined as
$$S_x(f_0) := \lim_{\Delta \rightarrow 0} \frac{1}{\Delta} P_y$$
- ❑ Can show this is equivalent to window definition
- ❑ Reveals how much power is in a certain frequency

# Spectrum Analyzer

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- ❑ Measures PSD in real time
- ❑ Uses averaging of FT
  - But proper averaging is quite tricky
- ❑ Lab 2: Use MATLAB function pwelch

# Units

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## □ Energy signals:

- $E_x$ : Joules
- $G(f) = |X(f)|^2$ : Joules / Hz

## □ Power signals (much more common):

- $P_x$ : Joules / sec = Watts
- $S_x(f)$ : Watts / Hz = Joules



# Power: Linear and decibel scale

---

## □ Receive or transmit antenna energy per unit time

- Measured in Watts (W) or mW
- Power values in W or mW called *linear scale*
- Use notation  $P_{|W}$  or  $P_{|mW}$  when units need to be specified

## □ Power often measured in dB scale:

- $P_{|dBW} = 10\log_{10}(P_{|W} / 1W)$
- $P_{|dBm} = 10\log_{10}(P_{|mW} / 1mW)$

## □ Example: $P = 250 \text{ mW}$ (typical max mobile transmit power)

- $P_{|dBW} = 10\log_{10}(0.25W / 1W)$
- $P_{|dBm} = 10\log_{10}(250mW / 1mW)$

# Some important dB values

---

□ Some conversions don't need a calculator:

- $10\log_{10}(2) = 3$  [Most important: Doubling power = 3dB]
- $10\log_{10}(3) = 4.7 \sim 5$
- $10\log_{10}(10) = 10$

□ You can cascade these.

□ Ex: If the power is increased by 50 in linear scale, what is the increase in dB? Answer:

$$\begin{aligned} 10\log_{10}(50) &= 10\log_{10}(10^2 / 2) \\ &= 2 \times 10\log_{10}(10) - 10\log_{10}(2) = 2 \times 10 - 3 = 17 \text{ dB} \end{aligned}$$

# PSD and Linear Filters

---

□ Suppose  $y(t) = h(t) * x(t)$

□ Then:  $Y(f) = H(f)X(f)$

- $S_y(f) = |H(f)|^2 S_x(f)$

□ Transfer function  $|H(f)|^2$  power gain at frequency  $f$ :

$$|H(f)|^2 = \frac{S_y(f)}{S_x(f)} \frac{\text{output power at } f}{\text{input power at } f}$$

- Dimensionless quantity
- Often expressed in dB

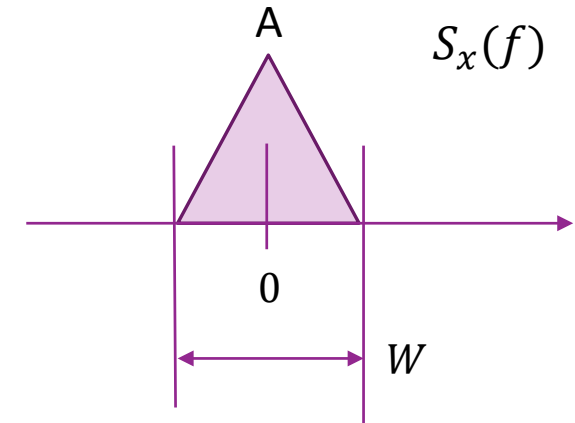
# Typical Wireless Power Transmit Levels

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- ❑ 100 kW = 80 dBm: Typical FM radio transmission with 50 km radius
- ❑ 1 kW = 60 dBm: Microwave oven element (most of this doesn't escape)
- ❑ ~300 W = 55 dBm: Geostationary satellite
- ❑ 250 mW = 24 dBm: Cellular phone maximum power (class 2)
- ❑ 200 mW = 23 dBm: WiFi access point
- ❑ 32 mW = 15 dBm: WiFi transmitter in a laptop
- ❑ 4 mW = 6 dBm: Bluetooth 10 m range
- ❑ 1 mW = 0 dBm: Bluetooth, 1 m range

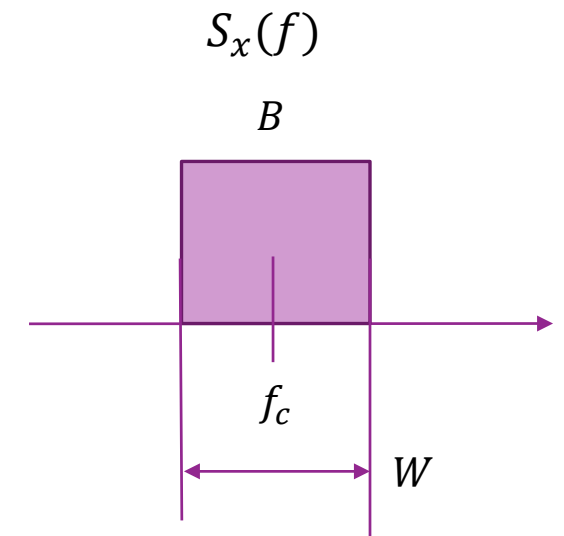
# Example 1:

- $S_x(f)$  is as shown.
- What is the power (in linear scale) in terms of  $W, A$ ?
- Suppose the power is  $P_x = 20$  dBm,  $W = 20$  MHz,  $f_c = 2$  GHz
  - What is  $A$ ?
  - What are the units of  $A$ ?
  
- Answer: on board



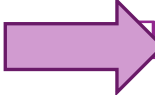
# Example 2

- $S_x(f)$  is shown for  $f > 0$ . Assume  $x(t)$  is real.  $W = 20$  MHz,  $B = 2(10)^{-8}$  mW/Hz,  $f_c = 2$  GHz
- What is  $P_x$ ? (Linear and in dBm).
- Suppose  $y(t) = h(t) * x(t)$  with  $H(f) = f_0 / (2\pi jf + f_0)$
- What is  $S_y(f)$ ? Draw it.
- Assuming  $f_c \gg f_0$  what is  $P_y$ ?
- What is the attenuation in dB?

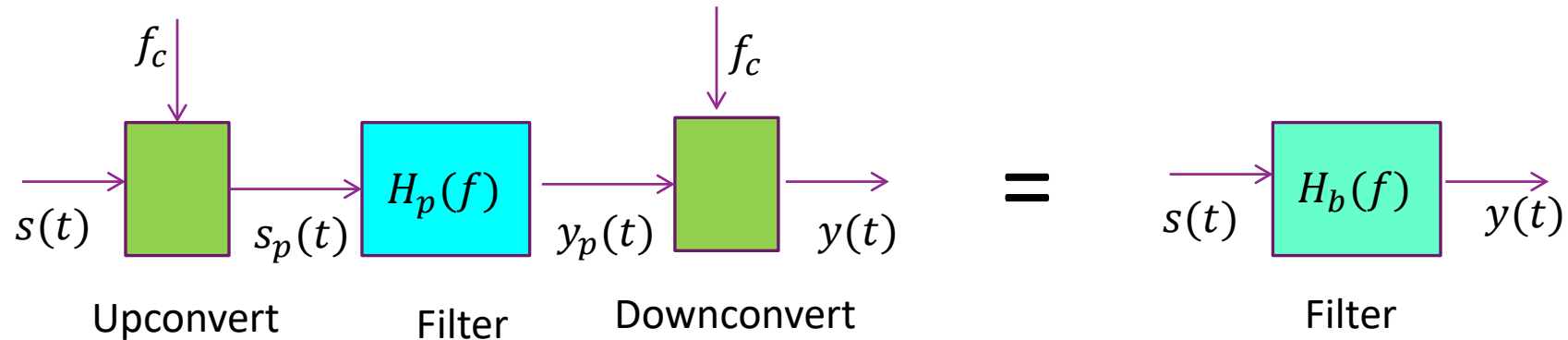


# Outline

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- ☐ Time-Domain Relationships
- ☐ Fourier Transform Review
- ☐ Frequency-Domain Relationships
- ☐ Power and Energy Spectra
-  ☐ Baseband equivalent filters
- ☐ Wireless channels

# Filtering



□ Filtering at passband equivalent to complex baseband filter

□ Assuming downconversion filter is ideal (see next slide):

- $H_b(f) = H_p(f + f_c)$  for  $|f| \leq \frac{W}{2}$
- Simply shift  $H_p(f)$  to the left by  $f_c$ .



# Proof of Result

---

□ Using the conversions from passband:

□ Downconversion formula:  $Y(f) = 2Y_p(f + f_c)H_{LPF}(f)$

□ Filtering in passband:  $Y(f) = 2H(f + f_c)U_p(f + f_c)H_{LPF}(f)$

□ Using upconversion formula:

$$Y(f) = H(f + f_c)\{U^*(-f - 2f_c) + U(f)\}H_{LPF}(f)$$

□ Assume:

- $U(f)H_{LPF}(f) \approx U(f)$  Filtering preserves baseband image
- $U^*(-f - 2f_c)H_{LPF}(f) \approx 0$  Filtering removes image around  $-2f_c$

□ Then  $Y(f) = H(f + f_c)U(f)$

# Delay

---

❑ Important special case: Suppose that  $h_p(t) = A\delta(t - \tau)$

- $A$  = gain
- $\tau$  = delay

❑ Passband frequency response is:  $H_p(f) = Ae^{-2\pi jf\tau}$

❑ Baseband frequency response:

$$H_b(f) = H_p(f + f_c) = Ae^{-2\pi j(f_c + f)\tau}$$

❑ Equivalent impulse response:

$$h_b(t) = Ae^{-2\pi jf_c\tau}\delta(t - \tau)$$

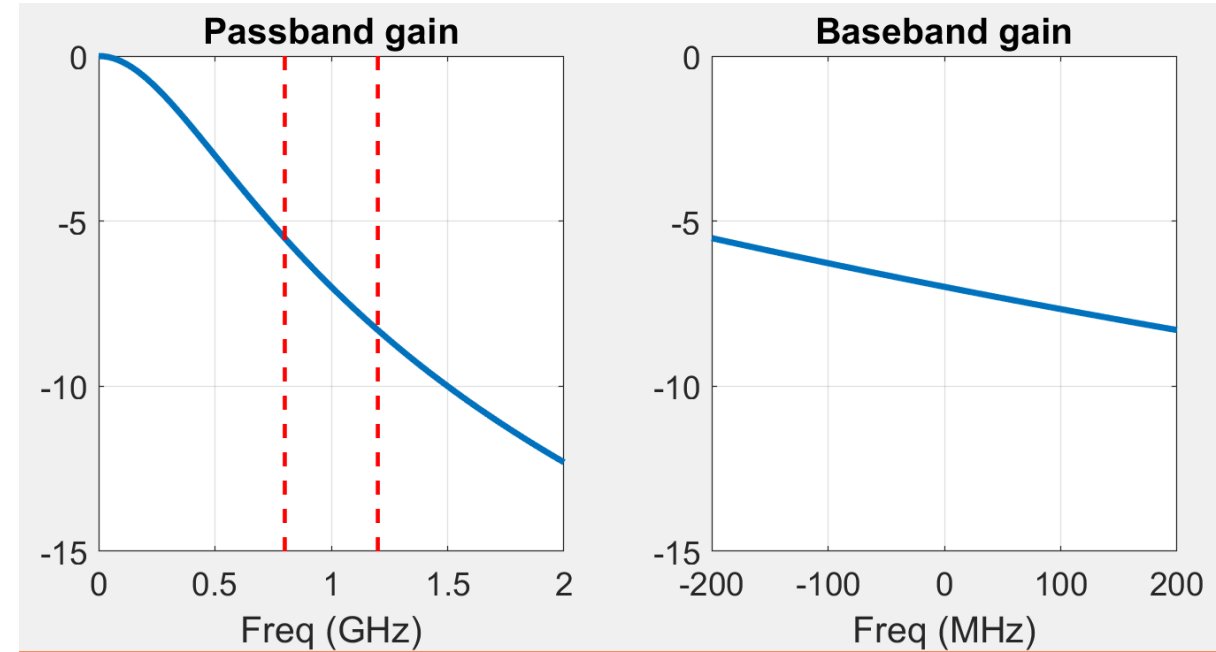
❑ Delay adds a constant phase rotation

# Example: First Order Filter

- Passband:  $H_p(\omega) = \frac{1}{1+j\omega/\omega_0}$
- Effective baseband:  $H(\omega) = \frac{1}{1+j(\omega+\omega_c)/\omega_0}$
- Observe baseband response is:
  - Almost flat
  - Not symmetric around  $f = 0$

```
fp = 1e9*linspace(0,2,128)';  
Hp = freqs(G0,[1/w0 1], 2*pi*fp);  
plot(fp/1e9, 20*log10(abs(Hp)), 'Linewidth', 3);
```

```
fb = linspace(-2e8,2e8,128)';  
Hb = freqs(G0, [1/w0, 1+1i*fc/f0], 2*pi*fb);  
plot(fb/1e6, 20*log10(abs(Hb)), 'Linewidth', 3);
```



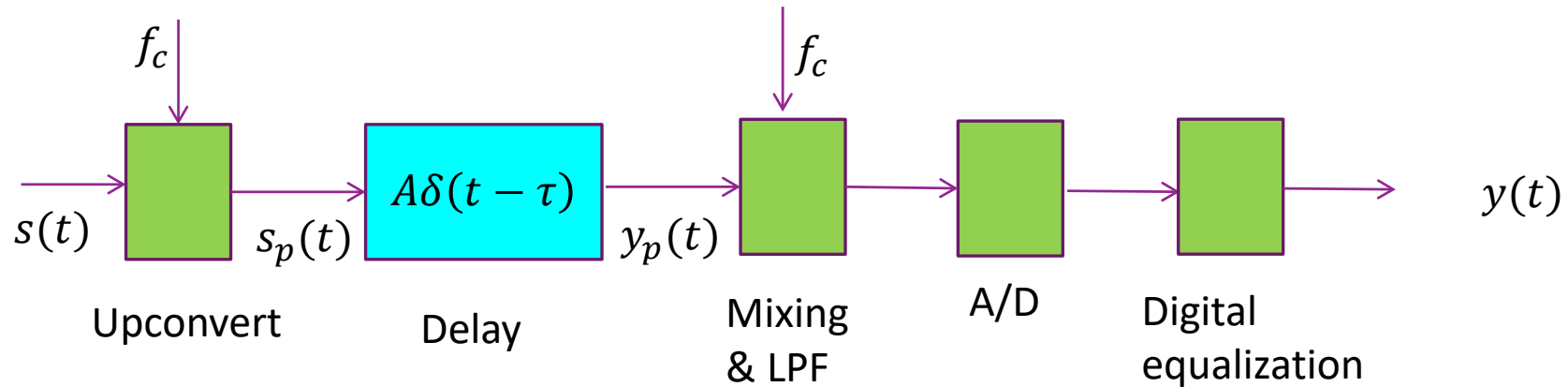
Passband cutoff freq  $f_0 = 0.5$  GHz  
Carrier freq  $f_c = 1$  GHz

# Frequency Errors

---

- ❑ LO on TX and RX are often slightly mismatched
- ❑ Suppose
  - Upconversion:  $u_p(t) = \text{Re}(u_{TX}(t)e^{j\omega_0 t + \theta_0})$
  - Downconversion:  $u_{RX}(t) = 2u_p(t)e^{-(j\omega_1 t + \theta_1)} + \text{LPF}$
- ❑ Then:
  - $u_{RX}(t) = u_{TX}(t)e^{j((\omega_0 - \omega_1)t + (\theta_0 - \theta_1))}$
- ❑ Causes a phase rotation

# Delay and Synchronization



- ❑ Two methods to compensate for delay at the RX
  - ❑ Method 1: Correct in analog by adjusting phase of LO
  - ❑ Method 2: Correct digital by inverting the gain  $Ae^{2\pi j f_c \tau}$ 
    - This is a special case of [equalization](#)

# Parts Per Million

---

- Oscillator error often measured in **parts per million** (ppm):

$$\Delta(\text{ppm}) := \frac{|f_c - f_c'|}{f_c} (10)^6$$

- $f_c$  = desired carrier frequency
- $f_c'$  = actual carrier frequency

- Example:

- $f_c = 2.5 \text{ GHz}$ ,  $\Delta = 10 \text{ ppm}$  (typ value for low-cost oscillator)
- Then,

$$|f_c - f_c'| = (2.5)(10)^9(10)(10)^6 = 25 \text{ kHz}$$

- Very large frequency shift!

# Outline

---

- ☐ Time-Domain Relationships
- ☐ Fourier Transform Review
- ☐ Frequency-Domain Relationships
- ☐ Power and Energy Spectra
- ☐ Baseband equivalent filters

 Wireless channels

# Freespace Wireless Channels

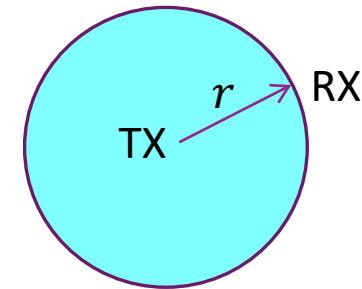
- ❑ Free space propagation
  - No obstacles
  - Isotropic (equal power in all directions)

- ❑ Power decreases as  $r^{-2}$ 
  - $\Rightarrow \text{Gain} = Ar^{-1}$  for some  $A$

- ❑ Delay is  $\tau = r/c$ ,  $c = 3(10)^8$  m/s

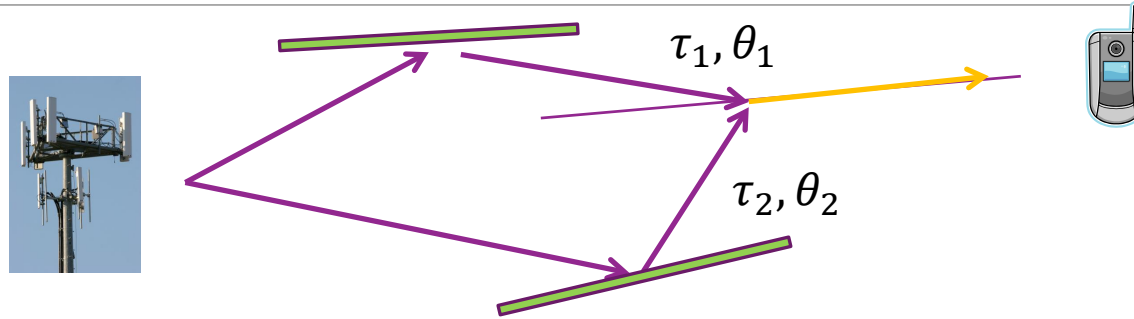
- ❑ Hence, baseband channel is:

$$h(t) = \frac{A}{r} e^{\frac{j2\pi f_c r}{c}} \delta(t - r/c)$$





# Multipath



❑ Wireless signals can arrive in many directions

- Reflections, diffraction, transmission, ...

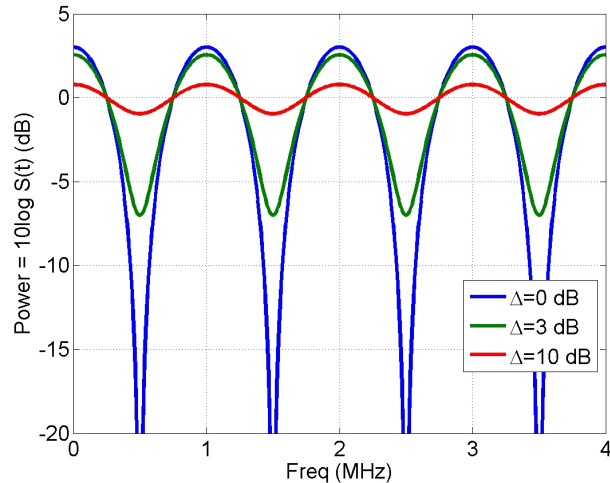
❑ Each path will have different gain and delay

❑ Receiver sees the combined total

$$h(t) = \sum_{k=1}^K h_k e^{-2\pi j f_c \tau_k} \delta(t - \tau_k)$$

- $h_k$  = complex gain of each path

# Two-Path Example



Magnitude response

$$S(f) = |H(f)|^2 = |h_1 e^{2\pi i f \tau_1} + h_2 e^{2\pi i f \tau_2}|^2$$

Plot shows:

$$\tau_2 - \tau_1 = 1 \mu s, |h_1|^2 + |h_2|^2 = 1, |h_2|^2 = 10^{0.1\Delta} |h_1|^2$$

□ Rate of variation in frequency depends on delay spread:  $\tau_2 - \tau_1$

□ Size of variation depends on spread of path gains:

- Average  $S(f) = |h_1|^2 + |h_2|^2$
- Min  $S(f) = (|h_1| - |h_2|)^2$ , Max  $S(f) = (|h_1| + |h_2|)^2$

# Example Problem (On board)

---

- ❑ A wireless channel has 2 paths:
  - Path 1: Power gain of -80 dB, travels 100m
  - Path 2: Power gain of -83 dB, travels 120m
- ❑ What are the amplitude gains of the two paths,  $h_1, h_2$ ?
- ❑ What are the two delays of the paths:  $\tau_1, \tau_2$ ?
- ❑ What is the average, minimum and maximum power?