

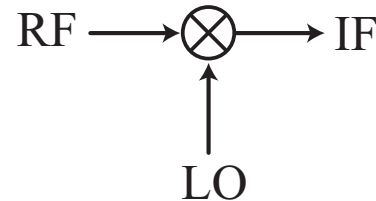
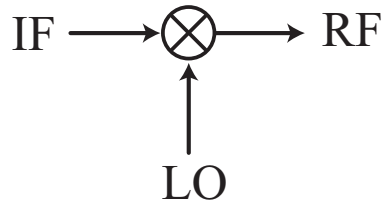
## *Lecture 15: Introduction to Mixers*

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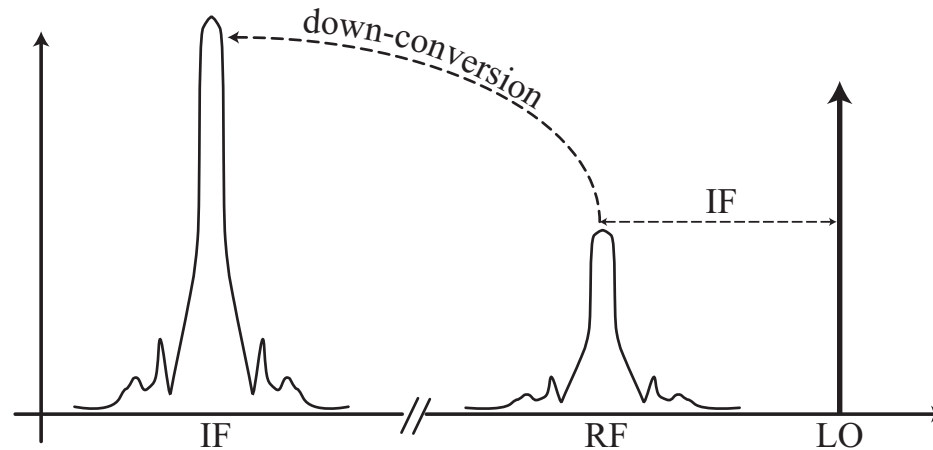
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# Mixers



- An ideal mixer is usually drawn with a multiplier symbol
- A real mixer cannot be driven by arbitrary inputs. Instead one port, the “LO” port, is driven by an *local oscillator* with a fixed amplitude sinusoid.
- In a *down-conversion* mixer, the other input port is driven by the “RF” signal, and the output is at a lower IF *intermediate frequency*
- In an *up-conversion* mixer, the other input is the IF signal and the output is the RF signal

# Frequency Translation



- As shown above, an ideal mixer translates the modulation around one carrier to another. In a receiver, this is usually from a higher RF frequency to a lower IF frequency. In a transmitter, it's the inverse.
- We know that an LTI circuit cannot perform frequency translation. Mixers can be realized with either time-varying circuits or non-linear circuits

# Ideal Multiplier

- Suppose that the input of the mixer is the RF and LO signal

$$v_{RF} = A(t) \cos(\omega_0 t + \phi(t))$$

$$v_{LO} = A_{LO} \cos(\omega_{LO} t)$$

- Recall the trigonometric identity

$$\cos(A + B) = \cos A \cos B - \sin A \sin B$$

- Applying the identity, we have

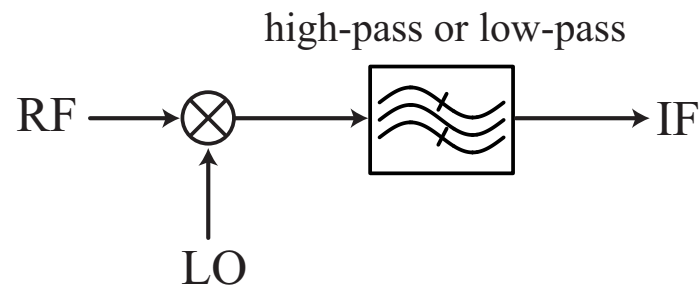
$$\begin{aligned} v_{out} &= v_{RF} \times v_{LO} \\ &= \frac{A(t)A_{LO}}{2} \{ \cos \phi (\cos(\omega_{LO} + \omega_0)t + \cos(\omega_{LO} - \omega_0)t) \\ &\quad - \sin \phi (\sin(\omega_{LO} + \omega_0)t + \sin(\omega_{LO} - \omega_0)t) \} \end{aligned}$$

# Ideal Multiplier (cont)

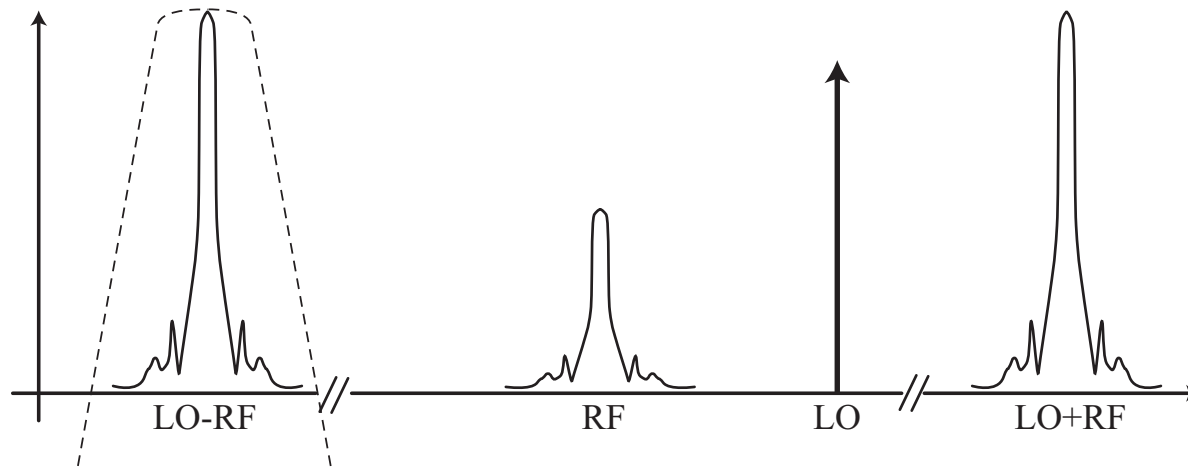
- Grouping terms we have

$$v_{out} = \frac{A(t)A_{LO}}{2} \left\{ \cos((\omega_{LO} + \omega_0)t + \phi(t)) + \cos((\omega_{LO} - \omega_0)t + \phi(t)) \right\}$$

- We see that the modulation is indeed translated to two new frequencies,  $LO + RF$  and  $LO - RF$ . We usually select either the upper or lower “sideband” by filtering the output of the mixer



# Mixer + Filter



- Note that the LO can be below the RF (lower side injection) or above the RF (high side injection)
- Also note that for a given LO, energy at  $LO \pm IF$  is converted to the same IF frequency. This is a potential problem!

# Upper/Lower Injection and Image

- Example: Downconversion Mixer

$$RF = 1\text{GHz} = 1000\text{MHz}$$

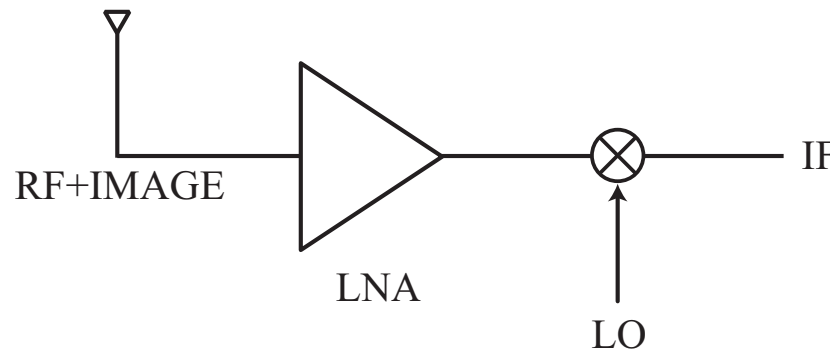
$$IF = 100\text{MHz}$$

Let's say we choose a low-side injection:

$$LO = 900\text{MHz}$$

That means that any signals or noise at 800MHz will also be downconverted to the same IF

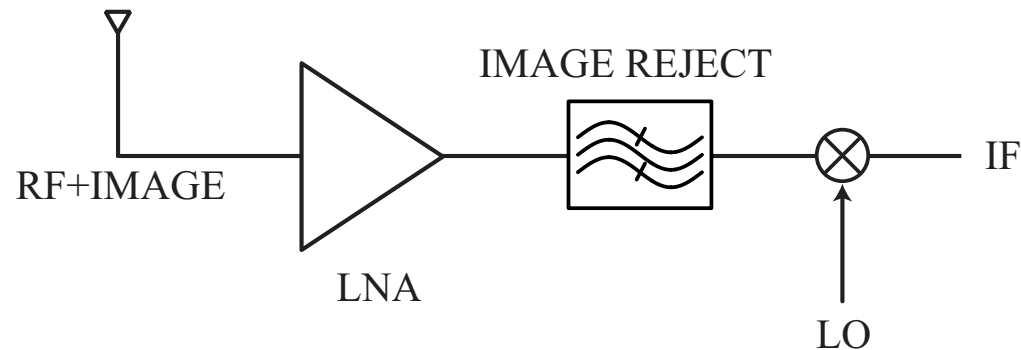
# Receiver Application



- The image frequency is the second frequency that also down-converts to the same IF. This is undesirable because the noise and interference at the image frequency can potentially overwhelm the receiver.
- One solution is to filter the image band. This places a restriction on the selection of the IF frequency due to the required filter  $Q$



# Image Rejection



- Suppose that  $RF = 1000\text{MHz}$ , and  $IF = 1\text{MHz}$ . Then the required filter bandwidth is much smaller than  $2\text{MHz}$  to knock down the image.
- In general, the filter  $Q$  is given by

$$Q = \frac{\omega_0}{BW} = \frac{RF}{BW}$$

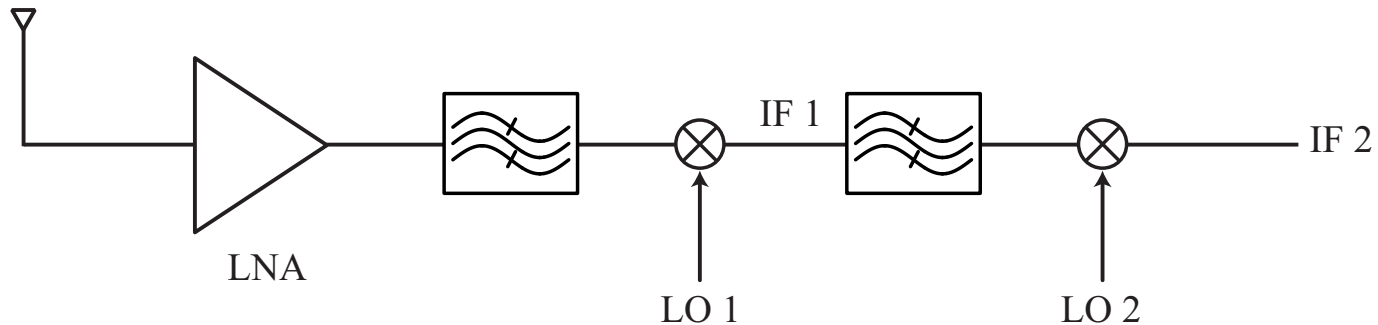
# Image Reject Filter

- In our example,  $RF = 1000\text{MHz}$ , and  $IF = 1\text{MHz}$ . The Image is on  $2IF = 2\text{MHz}$  away.
- Let's design a filter with  $f_0 = 1000\text{MHz}$  and  $f_1 = 1001\text{MHz}$ .
- A fifth-order Chebyshev filter with 0.2 dB ripple is down about 80 dB at the IF frequency.
- But the  $Q$  for such a filter is

$$Q = \frac{10^3\text{MHz}}{1\text{MHz}} = 10^3$$

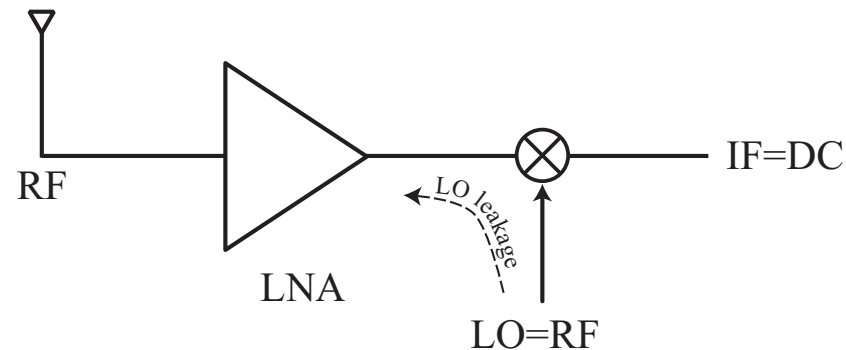
- Such a filter requires components with  $Q > 10^3!$

# RF Filtering



- The fact that the required filter  $Q$  is so high is related to the problem of filtering interferers. The very reason we choose the superheterodyne architecture is to simplify the filtering problem. It's much easier to filter a fixed IF than filter a variable RF.
- The image filtering problem can be relaxed by using multi-IF stages. Instead of moving to such a low IF where the image filtering is difficult (or expensive and bulky), we down-convert twice, using successively lower IF frequencies.

# Direct Conversion Receiver



- A mixer will frequency translate two frequencies,  $LO \pm IF$
- Why not simply down-convert directly to DC? Another words, why not pick a zero IF?
- This is the basis of the direct conversion architecture. There are some potential problems...

# Direction Conversion

- First, note that we must down-convert the desired signal and all the interfering signals. In other words, the LNA and mixer must be extremely linear.
- Since IF is at DC, all *even* order distortion now plagues the system, because the distortion at DC can easily swamp the desired signal.
- Furthermore, CMOS circuits produce a lot of flicker noise. Before we ignored this source of noise because it occurs at low frequency. Now it also competes with our signal.
- Another issue is with LO leakage. If any of the LO leaks into the RF path, then it will self-mix and produce a DC offset. The DC offset can rail the IF amplifier stages.

# Direct Conversion (cont)

- Example: If the IF amplifier has 80 dB of gain, and the mixer has 10 dB of gain, estimate the allowed LO leakage. Assume the ADC uses a 1V reference.
- To rail the output, we require a DC offset less than  $10^{-4}V$ . If the LO power is 0 dBm (316mV), we require an input leakage voltage  $< 10^{-5}V$ , or an isolation better than 90 dB!

# Phase of LO

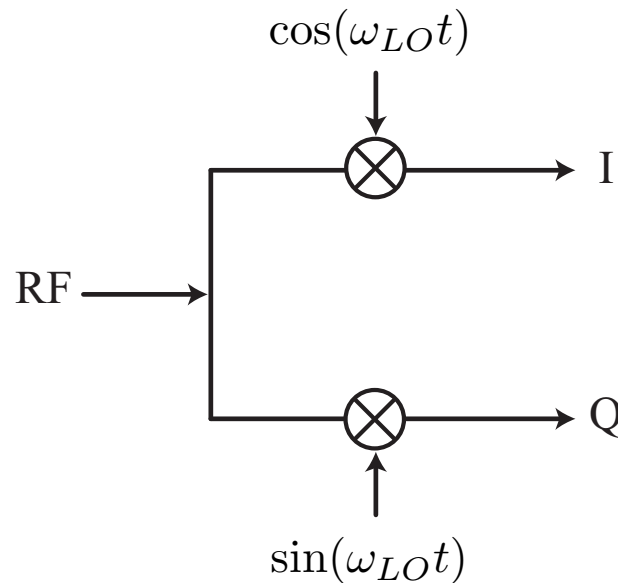
- In a direction conversion system, the LO frequency is equal to the RF frequency.
- Consider an input voltage  $v(t) = A(t) \cos(\omega_0 t)$ . Since the LO is generated “locally”, it’s phase is random relative to the RF input:

$$v_{LO} = A_{LO} \cos(\omega_0 t + \phi_0)$$

- If we are so unlucky that  $\phi_0 = 90^\circ$ , then the output of the mixer will be zero

$$\int_T A(t) A_{LO} \sin(\omega_0 t) \cos(\omega_0 t) dt$$
$$\approx A(t) A_{LO} \int_T \sin(\omega_0 t) \cos(\omega_0 t) dt = 0$$

# IQ-Mixer



- To avoid this situation, we can *phase lock* the LO to the RF by transmitting a pilot tone. Alternatively, we can use two mixers
- As we shall see, there are other benefits to such a mixer.



# AM Modulation

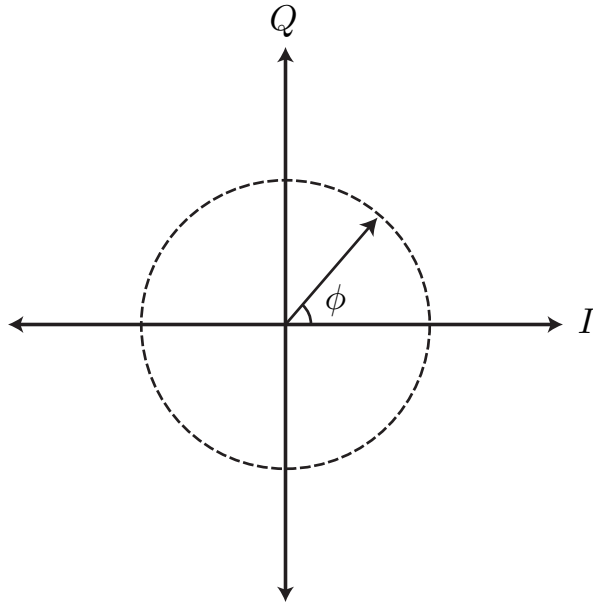
- We can see that an upconversion mixer is a natural amplitude modulator
- If the input to the mixer is a baseband signal  $A(t)$ , then the output is an AM carrier

$$v_o(t) = A(t) \cos(\omega_{LO}t)$$

- How do we modulate the phase? A PLL is one way to do it. The IQ mixer is another way.
- Let's expand a sinusoid that has AM and PM

$$\begin{aligned} v_o(t) &= A(t) \cos(\omega_0 t + \phi(t)) \\ &= A(t) \cos \omega_0 t \cos \phi(t) + A(t) \sin \omega_0 t \sin \phi(t) \\ &= I(t) \cos \omega_0 t + Q(t) \sin \omega_0 t \end{aligned}$$

# *I-Q* Plane



$$I(t) = A(t) \cos \phi(t)$$

$$Q(t) = A(t) \sin \phi(t)$$

- We can draw a trajectory of points on the *I-Q* plane to represent different modulation schemes.
- The amplitude modulation is given by

$$I^2(t) + Q^2(t) = A^2(t)(\cos^2 \phi(t) + \sin^2 \phi(t)) = A^2(t)$$

# General Modulator

- The phase modulation is given by

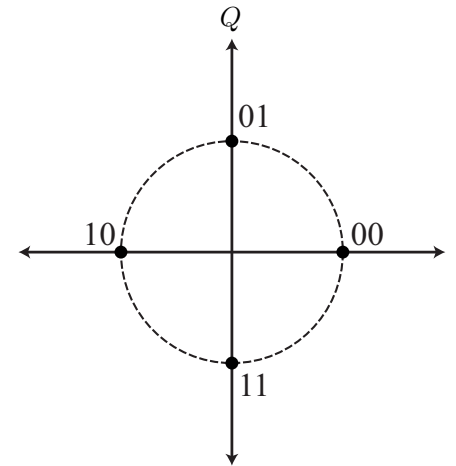
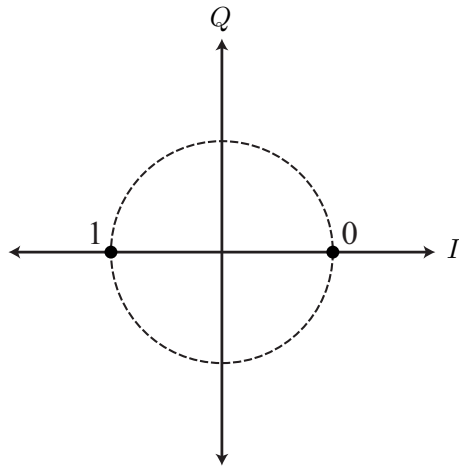
$$\frac{Q(t)}{I(t)} = \frac{\sin \phi(t)}{\cos \phi(t)} = \tan \phi(t)$$

or

$$\phi(t) = \tan^{-1} \frac{Q(t)}{I(t)}$$

- The IQ modulator is a universal digital modulator. We can draw a set of points in the IQ plane that represent symbols to transmit. For instance, if we transmit  $I = 0/A$  and  $Q = 0$ , then we have a simple ASK system (amplitude shift keying).

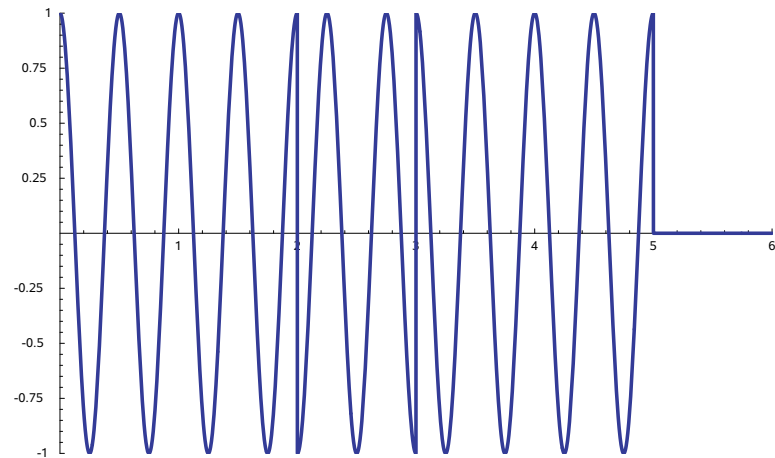
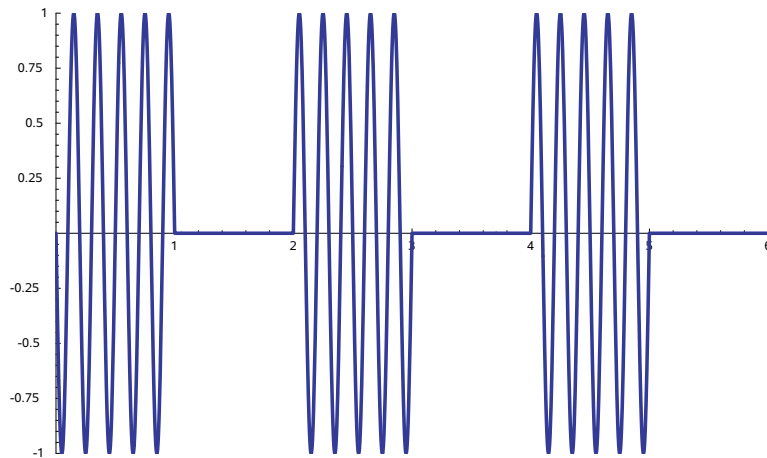
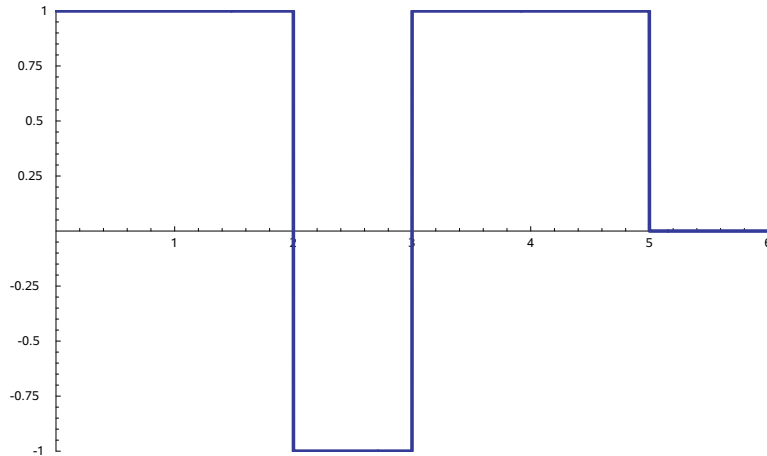
# Digital Modulation: BPSK/QPSK



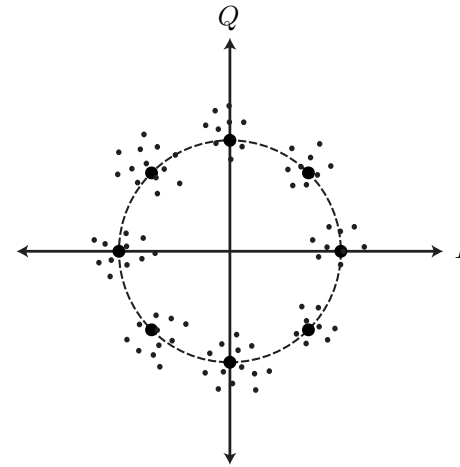
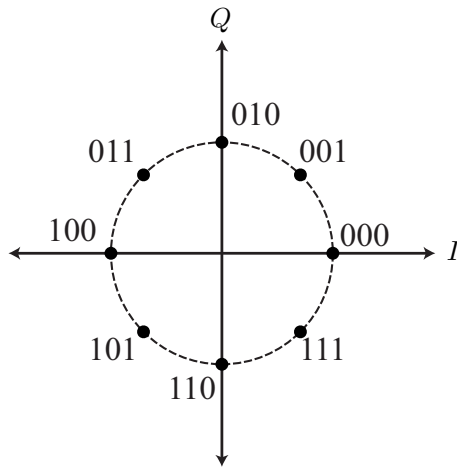
- For instance, if we transmit  $I(t) = \pm 1$ , this represents one bit transmission per cycle. But since the  $I$  and  $Q$  are orthogonal signals, we can improve the efficiency of transmission by also transmitting symbols on the  $Q$  axis.
- If we select four points on a circle to represent 2 bits of information, then we have a constant envelope modulation scheme.

# Modulation Waveforms

- The data wave form (left) is modulated onto a carrier (below). The first plot shows a simple on/off keying. The second plot shows binary phase shift keying on one channel (I).



# More Bits Per Cycle



- Eventually, the *constellation* points get very close together. Because of noise and distortion, the received spectrum will not lie exactly on the constellation points, but instead they will form a cluster around such points.
- If the clusters run into each other, errors will occur in the transmission.
- We can increase the radius but that requires more power.