

Integrated Circuits for Communication



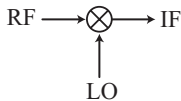
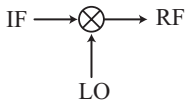
Berkeley

Introduction to Receivers and Mixers

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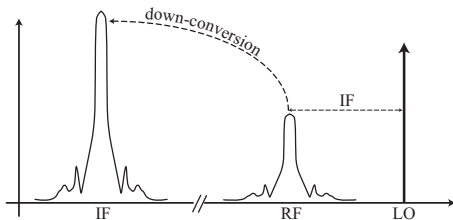
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- An ideal mixer is usually drawn with a multiplier symbol
- A real mixer cannot be driven by arbitrary inputs. Instead one port, the “LO” port, is driven by an *local oscillator* with a fixed amplitude sinusoid.
- In a *down-conversion* mixer, the 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 Down-Conversion



- 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.
- We know that an LTI circuit cannot perform frequency translation. Mixers can be realized with either time-varying circuits or non-linear circuits

- 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

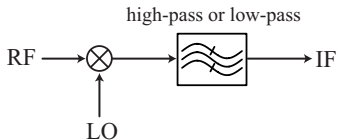
$$\begin{aligned} v_{out} &= v_{RF} \times v_{LO} \\ &= \frac{A(t)A_{LO}}{2} \{ \cos \phi (\cos(\omega_{LO} + \omega_0)t + \cos(\omega_{LO} - \omega_0)t) \\ &\quad - \sin \phi (\sin(\omega_{LO} + \omega_0)t + \sin(\omega_{LO} - \omega_0)t) \} \end{aligned}$$

Ideal Multiplier (cont)

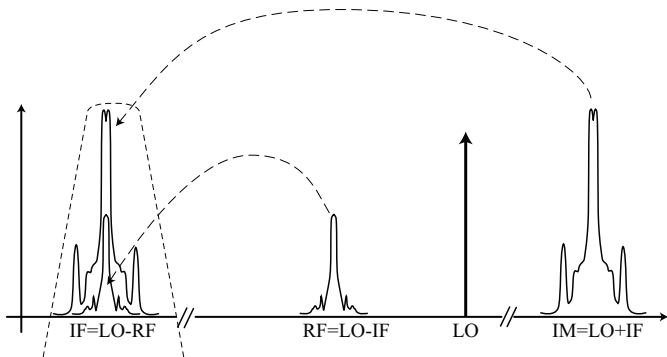
- Grouping terms we have

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

- 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!

- Example: Downconversion Mixer

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

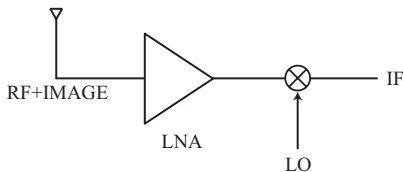
$$IF = 100\text{MHz}$$

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

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

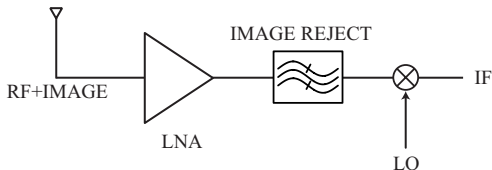
That means that any signals or noise at 800MHz will also be downconverted to the same IF

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 2MHz to knock down the image.
- In general, the filter Q is given by

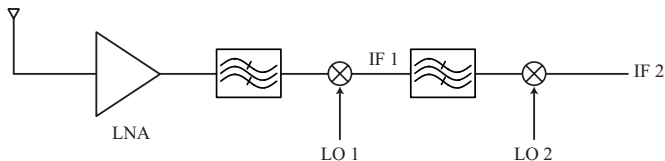
$$Q = \frac{\omega_0}{BW} = \frac{RF}{BW}$$

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

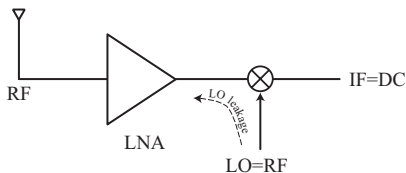
$$Q = \frac{10^3\text{MHz}}{1\text{MHz}} = 10^3$$

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



- 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? In other 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.
- Finally, if the modulation is complex, a simple mixer will garble the upper and lower side-band, a point we'll cover soon.

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!
- A better solution is to high-pass filter (if the modulation format allows it) or to cancel the offset voltage with a DAC in the baseband.

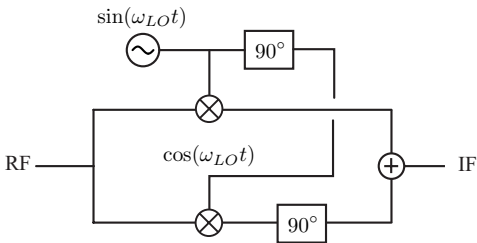
- In a direction conversion system, the LO frequency is equal to the RF frequency.
- Consider an input voltage $v(t) = A(t) \cos(\omega_0 t)$. Since the LO is generated “locally”, it’s phase is random relative to the RF input:

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

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

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

I/Q Hartley Mixer



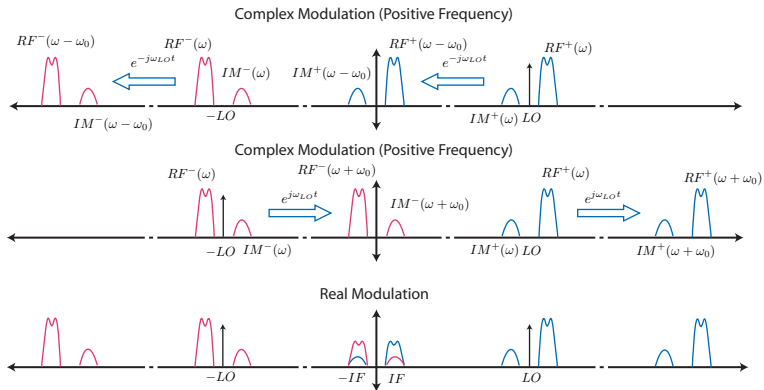
- An I/Q mixer implemented as shown above is known as a Hartley Mixer.
- We will also show that such a mixer can perform image rejection.

- Consider the action of a 90° delay on an arbitrary signal. Clearly $\sin(x - 90^\circ) = -\cos(x)$. Even though this is obvious, consider the effect on the complex exponentials

$$\begin{aligned}\sin\left(x - \frac{\pi}{2}\right) &= \frac{e^{jx-j\pi/2} - e^{-jx+j\pi/2}}{2j} \\ &= \frac{e^{jx}e^{-j\pi/2} - e^{-jx}e^{j\pi/2}}{2j} = \frac{e^{jx}(-j) - e^{-jx}(j)}{2j} \\ &= -\frac{e^{jx} + e^{-jx}}{2} = -\cos(x)\end{aligned}$$

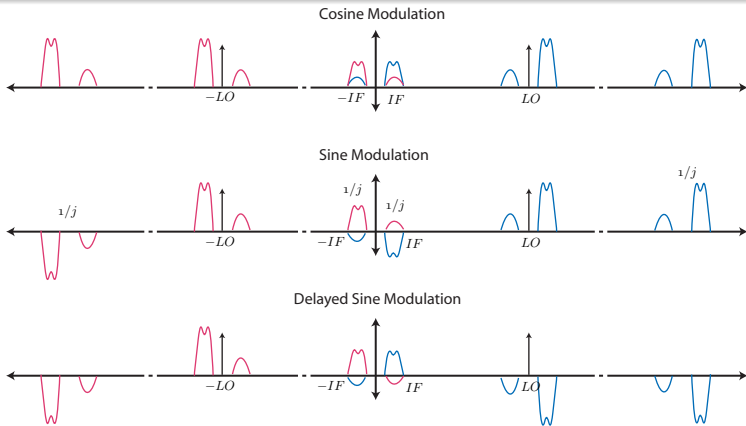
- Positive frequencies get multiplied by $-j$ and negative frequencies by $+j$. This is true for a narrowband signal when it is delayed by 90° .

Image Problem (Again)



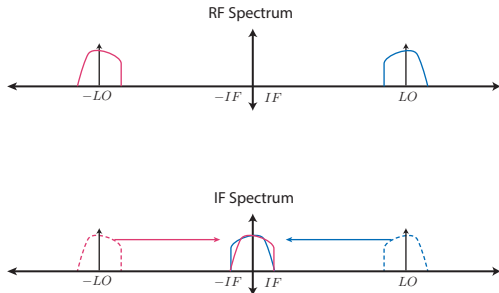
- We see that the image problem is due to multiplication by the sinusoid and not a complex exponential. If we could synthesize a complex exponential, we would not have the image problem.

Sine/Cosine Modulation



- Using the same approach, we can find the result of multiplying by \sin and \cos as shown above. If we delay the \sin portion, we have a very desirable situation! The image is inverted with respect to the \cos and can be cancelled.

Direct Down-Conversion with Complex Modulated Waveform



- Note that if the signal is a complex modulated signal up-converted, then if we simply downconvert it with a tone (sin or cos), the image reject problem get translated into a spectrum mangling problem.
- For this reason, a complex down-converter is required, which explains why most modern communication systems use a complex I and Q mixer.

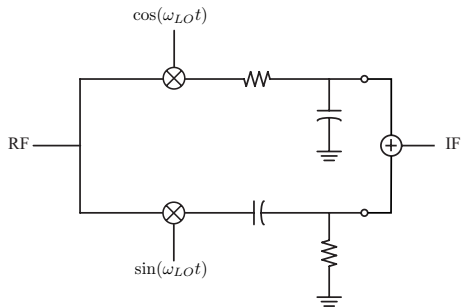
Image Rejection Matching Requirements

- The image rejection scheme just described is very sensitive to phase and gain match in the I/Q paths. Any mismatch will produce only finite image rejection.
- The image rejection for a given gain/phase match is approximately given by

$$IRR(\text{ dB}) = 10 \cdot \log \frac{1}{4} \left(\left(\frac{\delta A}{A} \right)^2 + (\delta\theta)^2 \right)$$

- For typical gain mismatch of 0.2 – 0.5 dB and phase mismatch of $1^\circ - 4^\circ$, the image rejection is about 30 dB - 40 dB. We usually need about 60 – 70 dB of total image rejection.

$\pm 45^\circ$ Delay Element



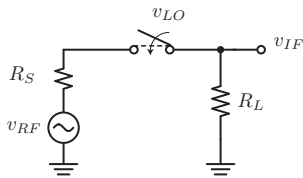
- The passive R/C and C/R lowpass and highpass filters are a nice way to implement the delay. Note that their relative phase difference is always 90° .

$$\angle H_{lp} = \angle \frac{1}{1 + j\omega RC} = -\arctan \omega RC$$

$$\angle H_{hp} = \angle \frac{j\omega RC}{1 + j\omega RC} = \frac{\pi}{2} - \arctan \omega RC$$

- But to have equal gain, the circuit must operate at the $1/RC$ frequency. This restricts the circuit to relatively narrowband systems. Multi-stage polyphase circuits remedy the situation but add insertion loss to the circuit.
- The I/Q LO signal is usually generated directly rather than through an high-pass and low-pass network.
- Two ways to generate the I/Q LO is through a divide-by-two circuit (requires $2 \times LO$) or a quadrature oscillator (requires two tanks).

Practical Mixer Realization



- Real mixers are realized not as “multipliers” but using switches. The above schematic is in the core building block for a mixer: a switch !
- The control port of the switch is driven by the periodic LO signal (square or sine wave), and hence the transfer function varies periodically:

$$v_{IF}(t) = v_{RF}(t) \frac{R_L}{R_L + R_{SW}(t)}$$

$$R_{SW}(t) = f(v_{LO}(t))$$

Mixer Analysis: Time Domain

- A generic mixer operates with a periodic transfer function $h(t + T) = h(t)$, where $T = 1/\omega_0$, or T is the LO period. We can thus expand $h(t)$ into a Fourier series

$$y(t) = h(t)x(t) = \sum_{-\infty}^{\infty} c_n e^{j\omega_0 n t} x(t)$$

- For a sinusoidal input, $x(t) = A(t) \cos \omega_1 t$, we have

$$y(t) = \sum_{-\infty}^{\infty} \frac{c_n}{2} A(t) \left(e^{j(\omega_1 + \omega_0 n)t} + e^{j(-\omega_1 + \omega_0 n)t} \right)$$

- Since $h(t)$ is a real function, the coefficients $c_{-k} = c_k$ are even. That means that we can pair positive and negative frequency components.

Time Domain Analysis (cont)

$$= c_1 \frac{e^{j(\omega_1 + \omega_0)t} + e^{j(-\omega_1 + \omega_0)t}}{2} A(t) + c_{-1} \frac{e^{j(\omega_1 - \omega_0)t} + e^{j(-\omega_1 - \omega_0)t}}{2} A(t) + \dots$$

- Take c_1 and c_{-1} as an example ($c_1 = c_{-1}$)

$$= c_1 A(t) \cos(\omega_1 + \omega_0)t + c_1 A(t) \cos(\omega_1 - \omega_0)t + \dots$$

- Summing together all the components, we have

$$y(t) = \sum_{-\infty}^{\infty} c_n \cos(\omega_1 + n\omega_0)t$$

- Unlike a perfect multiplier, we get an infinite number of frequency translations up and down by harmonics of ω_0 .

Frequency Domain Analysis

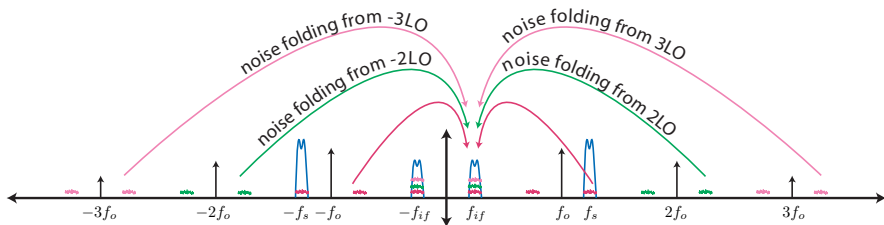
- Since multiplication in time, $y(t) = h(t) \cdot x(t)$, is convolution in the frequency domain, we have

$$Y(f) = H(f) * X(f)$$

- The transfer function $H(f) = \sum_{-\infty}^{\infty} c_n \delta(f - nf_0)$ has a discrete spectrum. So the output is given by

$$\begin{aligned} Y(f) &= \int_{-\infty}^{\infty} \sum_{-\infty}^{\infty} c_n \delta(\sigma - nf_0) X(f - \sigma) d\sigma \\ &= \sum_{-\infty}^{\infty} c_n \int_{-\infty}^{\infty} \delta(\sigma - nf_0) X(f - \sigma) d\sigma \end{aligned}$$

Frequency Domain (cont)



- By the frequency sifting property of the $\delta(f - \sigma)$ function, we have

$$Y(f) = \sum_{-\infty}^{\infty} c_n X(f - nf_0)$$

- Thus, the input spectrum is shifted by all harmonics of the LO up and down in frequency.

Noise/Image Problem

- Previously we examined the “image” problem. Any signal energy a distance of IF from the LO gets downconverted in a perfect multiplier. But now we see that for a general mixer, any signal energy with an IF of any harmonic of the LO will be downconverted !
- These other images are easy to reject because they are distant from the desired signal and a image reject filter will be able to attenuate them significantly.
- The noise power, though, in all image bands will fold onto the IF frequency. Note that the noise is generated by the mixer source resistance itself and has a white spectrum. Even though the noise of the antenna is filtered, new noise is generated by the filter itself!

Mixer Noise Definition

- By definition we have $F = \frac{SNR_i}{SNR_o}$. If we apply this to a receiving mixer, the input signal is at the “RF” and the output signal is at “IF”. There is some ambiguity to this definition because we have to specify if the RF signal is a single (upper or lower) or double sideband modulated waveform.
- For a single-sideband modulated waveform, the noise from the image band adds, therefore doubling the IF noise relative to RF. Thus the $F = 2$.
- For a double sideband modulated waveform, though, there is signal energy in both sidebands and so for a perfect multiplying mixer, $F = 1$ since the IF signal is twice as large since energy from both sidebands fall onto the IF.
- If an image reject filter is used, the noise in the image band can be suppressed and thus $F = 1$ for a cascade of a sharp image reject filter followed by a multiplier.