

# **Modulation. Analog Modulation and Demodulation**

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# Outline

The Concept of Modulation

Amplitude Modulation

Demodulation

# Generalized Fourier Transform

- Consider a DC or constant signal:

$$x(t) = 1, -\infty < t < \infty$$

- Compute its Fourier Transform:

$$X(j\omega) = \int_{-\infty}^{\infty} (1)e^{-j\omega t} dt = \lim_{T \rightarrow \infty} \int_{-T/2}^{T/2} e^{-j\omega t} dt = \lim_{T \rightarrow \infty} -\frac{1}{j\omega} e^{-j\omega t} \Big|_{-T/2}^{T/2} = \lim_{T \rightarrow \infty} -\frac{1}{j\omega} [e^{-j\omega T/2} - e^{j\omega T/2}]$$

- Unfortunately, the limit is not finite, and the integral does not converge.
- Consider an alternate approach based on an impulse function:

$$\delta(t) = 0 \quad t \neq 0, \quad \int_{-\varepsilon}^{\varepsilon} \delta(\lambda) d\lambda = 1, \quad \varepsilon > 0 \quad \leftrightarrow \quad \mathcal{F}\{\delta(t)\} = 1$$

- Apply the duality property:

$$x(t) = 1 \quad -\infty \leq t \leq \infty \quad \leftrightarrow \quad \mathcal{F}\{x(t)\} = 2\pi\delta(\omega)$$

- This is known as the **Generalized Fourier transform**. It allows us to extend the Fourier transform to some additional useful signals such as periodic signals:

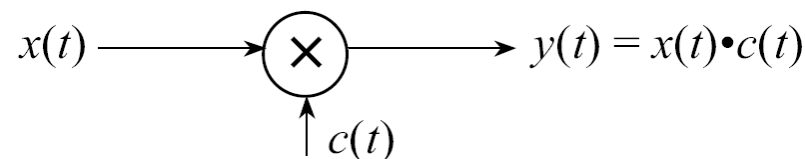
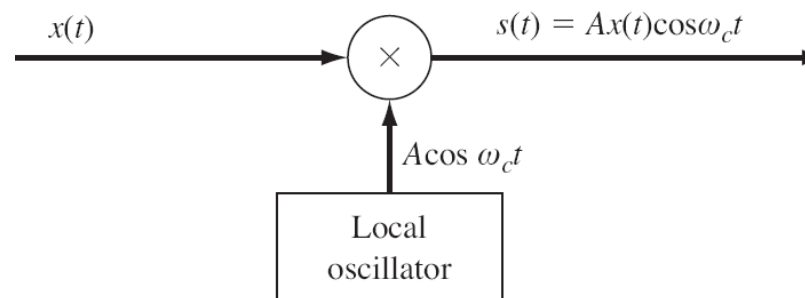
$$e^{j\omega_0 t} \leftrightarrow 2\pi\delta(\omega - \omega_0)$$

$$x(t) = \sum_{k=-\infty}^{\infty} c_k e^{jk\omega_0 t} \leftrightarrow X(\omega) = \sum_{k=-\infty}^{\infty} 2\pi c_k \delta(\omega - k\omega_0)$$

The Fourier transform of a periodic signal is a train of impulse functions (and is a line spectrum).

# The Concept of Modulation

- The electromagnetic spectrum is the most expensive “real estate” in the world. Hence, we would like to make as efficient use of it as possible (e.g., time and frequency domain multiplexing).
- It is more efficient (e.g., less power for a given SNR) to transmit signals at higher frequencies.
- **Modulation:** send multiple signals through the same medium (e.g. air, cables, fibers) by simply shifting them to different places in the spectrum.
- **Amplitude Modulation:** carry the information in the amplitude of the signal; use a sinusoidal carrier,  $c(t)$ .
- **Angle Modulation:** alternate approach in which the signal is carrier signal (frequency and phase modulation).
- Many other forms of modulation including pulse-amplitude modulation (PAM), pulse-width modulation (PWM), code division multiple access (CDMA) and spread spectrum. These techniques are typically studied in an introductory course in communications theory.



# Amplitude Modulation Using a Complex Exponential

- **Modulation:**

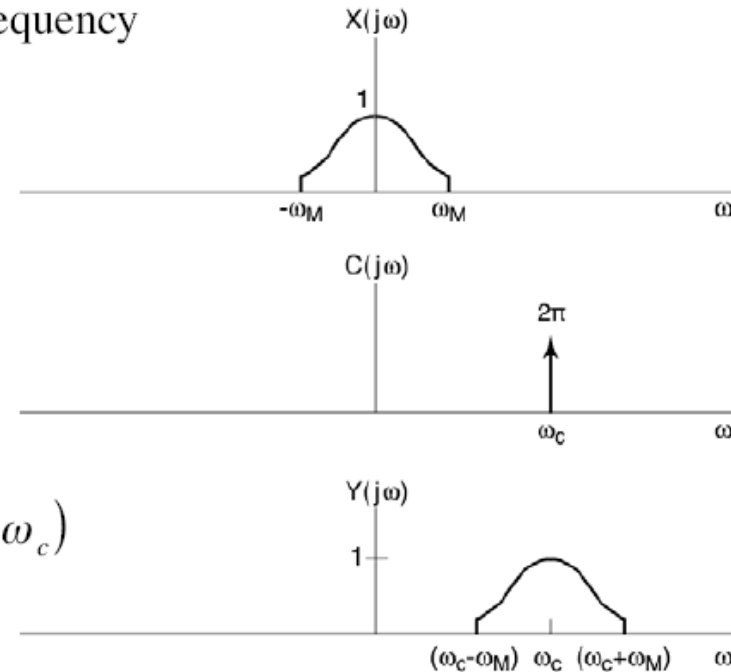
$$c(t) = e^{j\omega_c t}, \quad \omega_c \text{ — carrier frequency}$$

$$y(t) = x(t) e^{j\omega_c t}$$

$$Y(j\omega) = \frac{1}{2\pi} X(j\omega) * C(j\omega)$$

$$= \frac{1}{2\pi} X(j\omega) * 2\pi\delta(\omega - \omega_c)$$

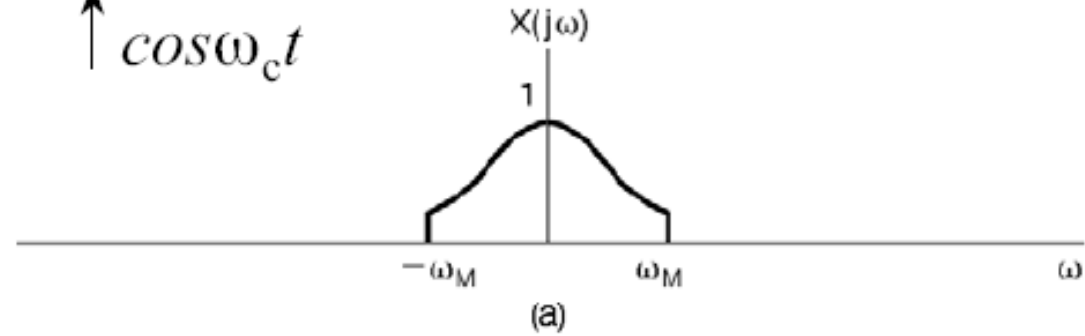
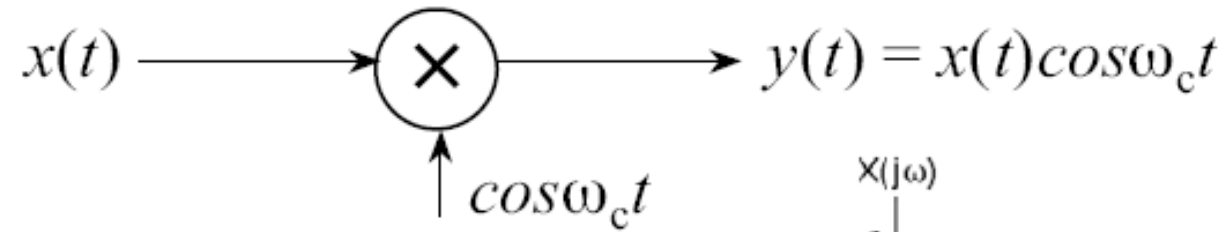
$$= X(j(\omega - \omega_c))$$



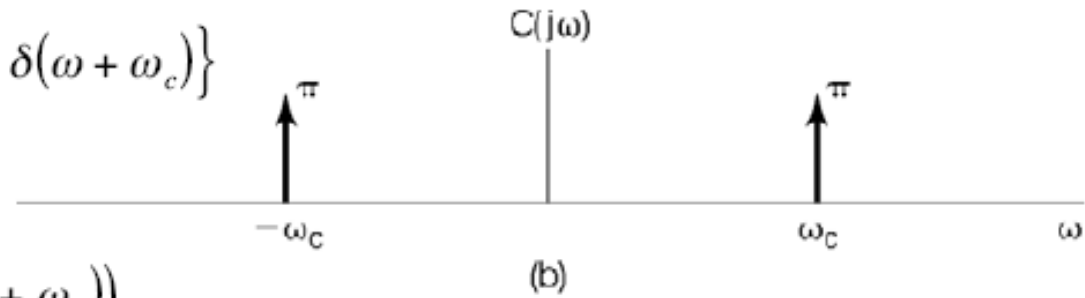
- **Demodulation:**



# Amplitude Modulation Using a Sinusoid



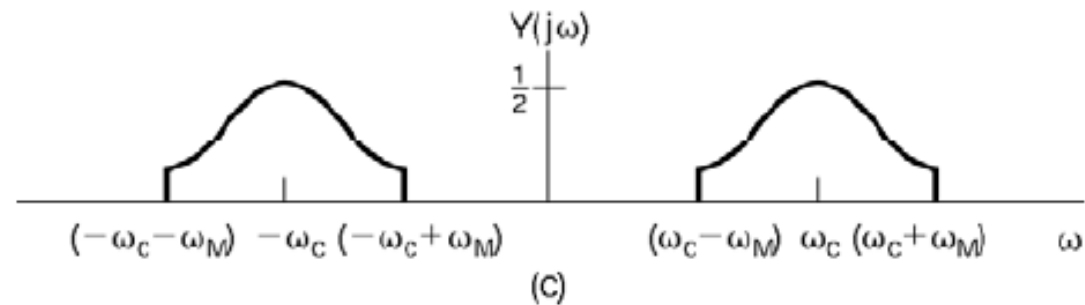
$$Y(j\omega) = \frac{1}{2\pi} X(j\omega) * \pi \{ \delta(\omega - \omega_c) + \delta(\omega + \omega_c) \}$$



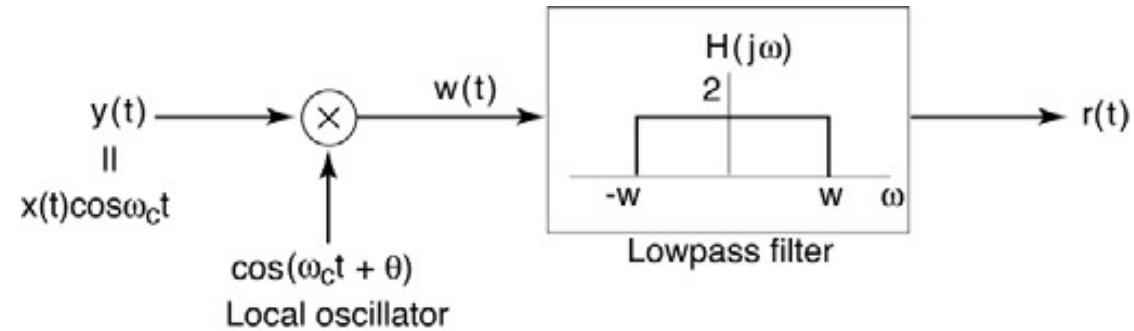
$$= \frac{1}{2} X(j(\omega - \omega_c)) + \frac{1}{2} X(j(\omega + \omega_c))$$

Drawn assume

$$\omega_c > \omega_M$$



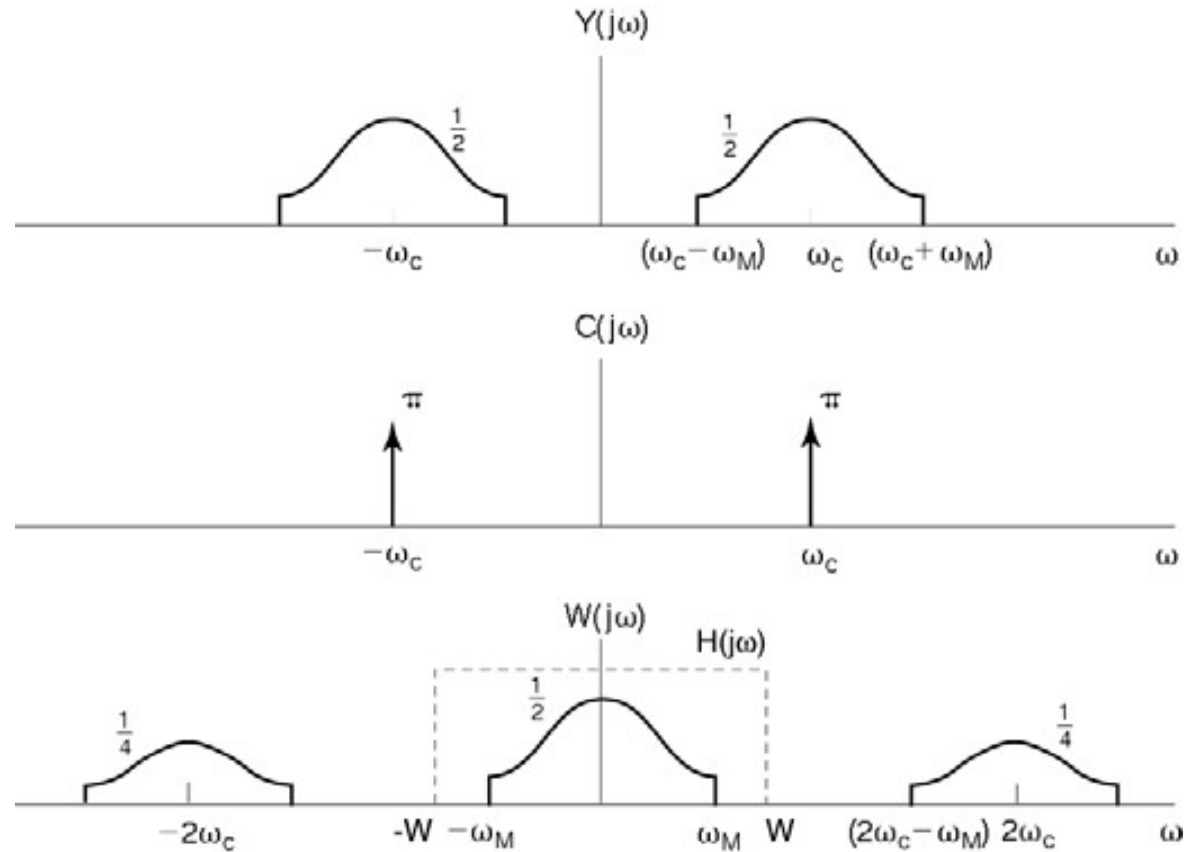
# Synchronous Demodulation of Sinusoidal AM



- **Assumptions:**

- $\theta = 0$  (for now),
- Local oscillator is *synchronized with the carrier*.

In practice, synchronization is achieved using a phase-locked loop (PLL).



# Synchronous Demodulation in the Time Domain

- We can easily derive the properties of the demodulated signal:

$$w(t) = y(t) \cos(\omega_c t) = x(t) \cos^2(\omega_c t) = x(t) \left[ \frac{1}{2} + \frac{1}{2} \cos(2\omega_c t) \right]$$

- The low-pass filter removes the high-frequency replica of  $x(t)$ , leaving only the “baseband” component.
- Suppose there is a phase difference between the transmitter and the receiver:

$$w(t) = y(t) \cos(\omega_c t + \theta) = x(t) \cos(\omega_c t) \cos(\omega_c t + \theta)$$

The mismatch in phase appears as a scale factor  $\left[ \frac{1}{2} \cos(\theta) + \frac{1}{2} \cos(2\omega_c t + \theta) \right]$  that can be ignored.

- If there is a time-varying phase difference (due to drift):

$$w(t) = x(t) \left[ \frac{1}{2} \cos(\theta(t)) + \frac{1}{2} \cos(2\omega_c t + \theta(t)) \right]$$

If the phase difference varies slowly in time, the net result is simply a time-varying amplitude change, which distorts the signal (slightly).

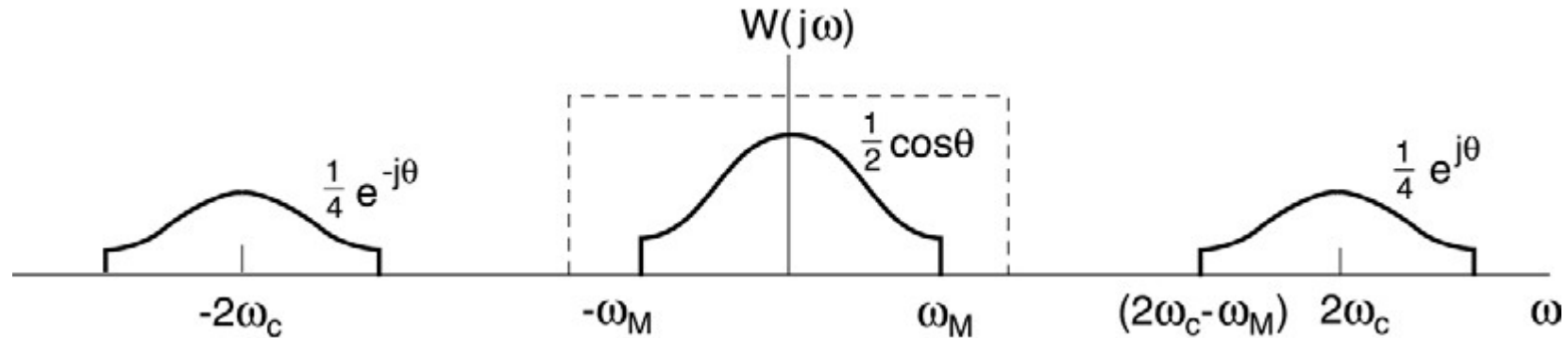
- What happens if the receiver is exactly  $90^\circ$  out of phase?



# Asynchronous Demodulation

- Consider the spectrum of our modulation signal:

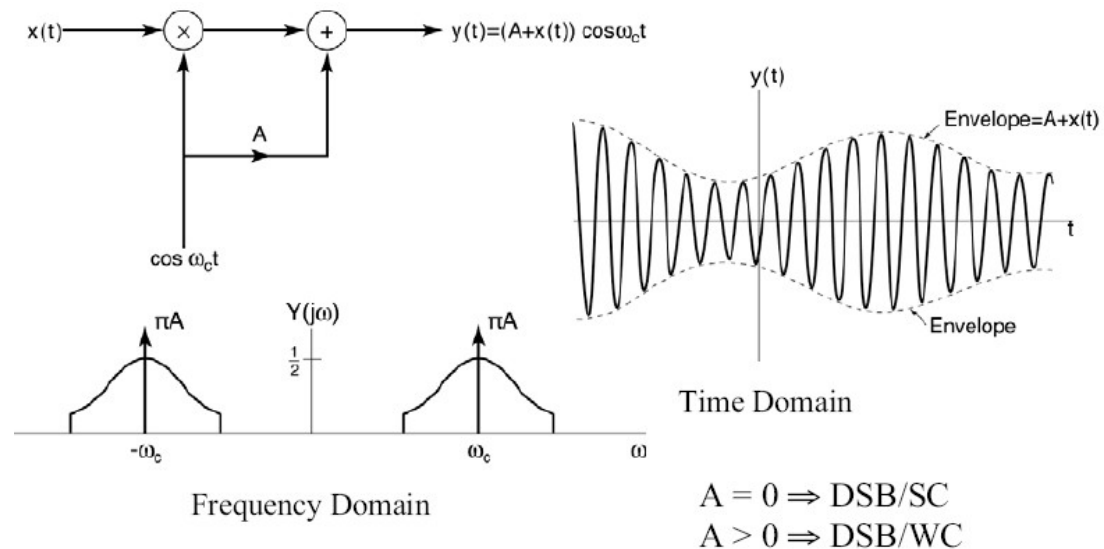
$$\cos(\omega_c t + \theta) = \frac{1}{2} e^{j\theta} e^{j\omega_c t} + \frac{1}{2} e^{-j\theta} e^{-j\omega_c t} \leftrightarrow \pi e^{j\theta} \delta(\omega - \omega_c) + \pi e^{-j\theta} \delta(\omega + \omega_c)$$



Problems if the transmitter and receiver are exactly  $90^\circ$  out of phase.

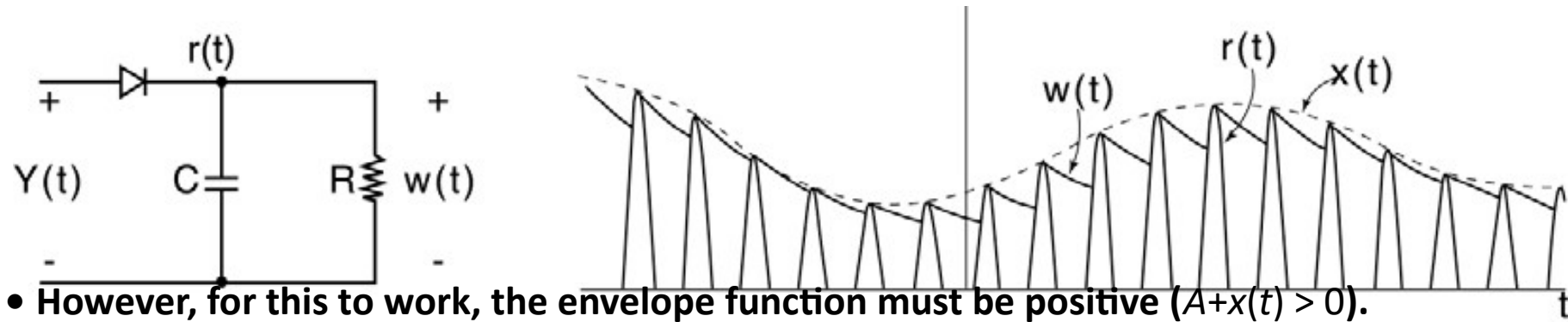
- Alternative: **Asynchronous modulation** includes the carrier in the output and ensures the envelope of the modulated carrier contains the information.

- The carrier must be much higher in frequency than the signal:  $\omega_c > \omega_M$



# Implementation of an Asynchronous Demodulator

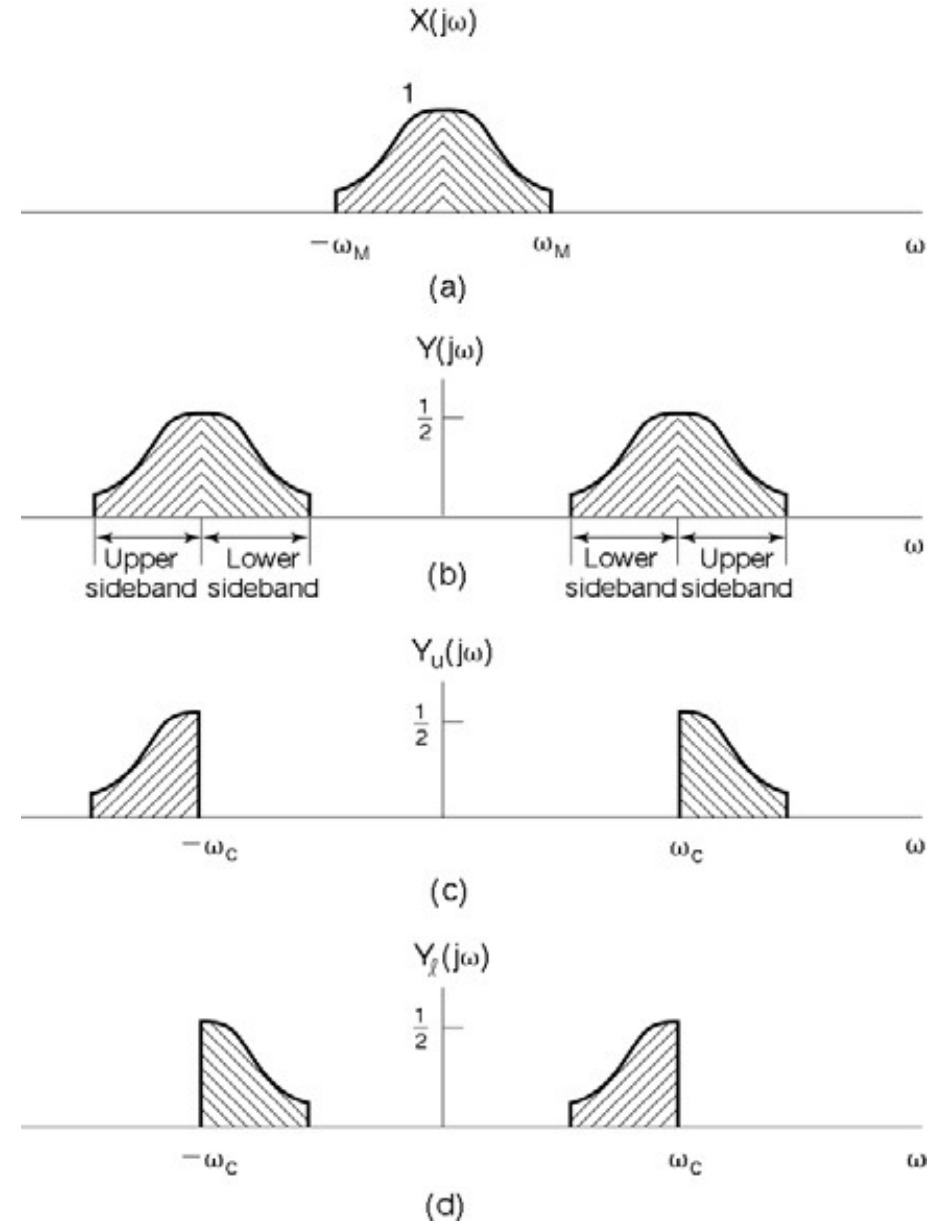
- AM modulation was popular because of the ease with which it could be demodulated:



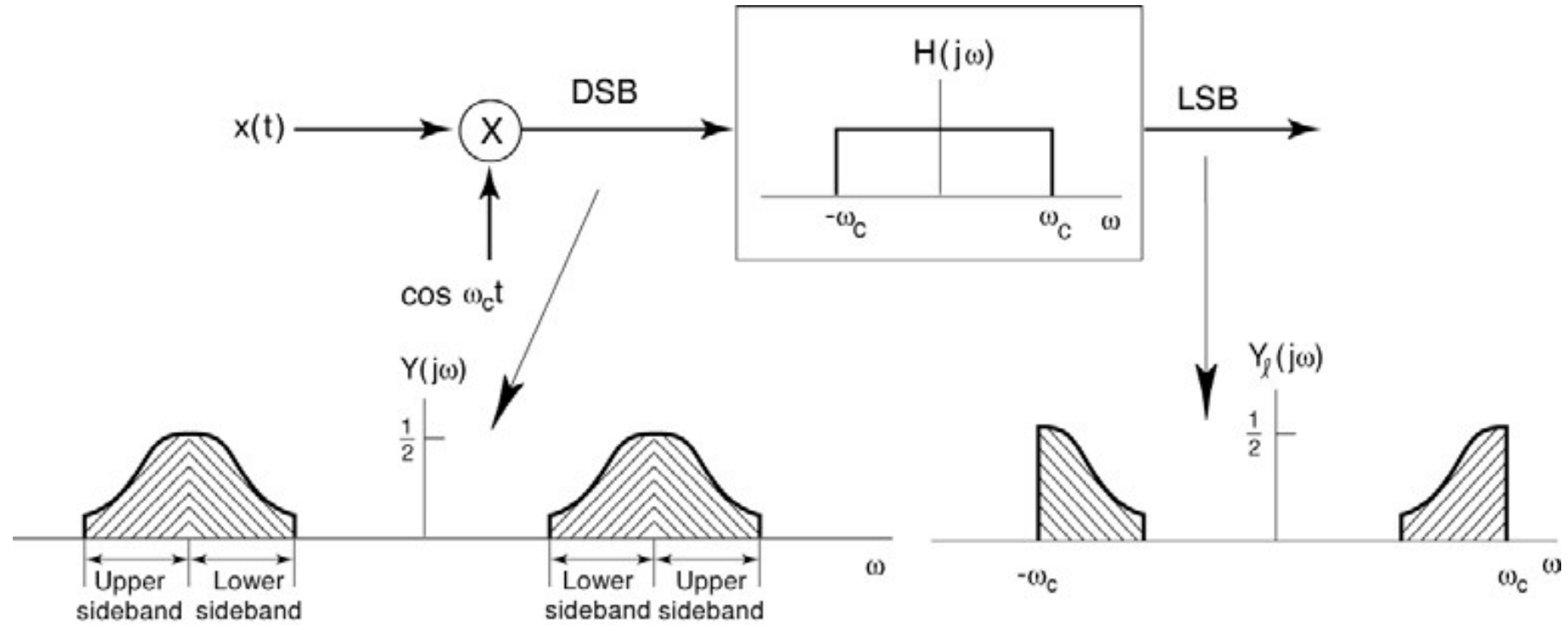
- However, for this to work, the envelope function must be positive ( $A+x(t) > 0$ ).
- We pay a price in efficiency: more power must be used to guarantee this condition is true.

# Double-Sideband Vs. Single-Sideband Modulation

- Since  $x(t)$  and  $y(t)$  are real, from conjugate symmetry, both lower sideband (LSB) and upper sideband (USB) signals carry exactly the same information.
- Double-sideband (DSB) occupies  $2M$  bandwidth in  $\omega > 0$ , even though *all* the information is contained in  $M$ .
- Single-sideband (SSB) occupies  $M$  bandwidth in  $\omega > 0$ .
- Of course, SSB requires slightly more complicated hardware, so it was originally only used in applications where bandwidth was very limited (e.g., transcontinental telephone lines).
- Analog television signals, which are being obsoleted in February 2009, use a variant of SSB.

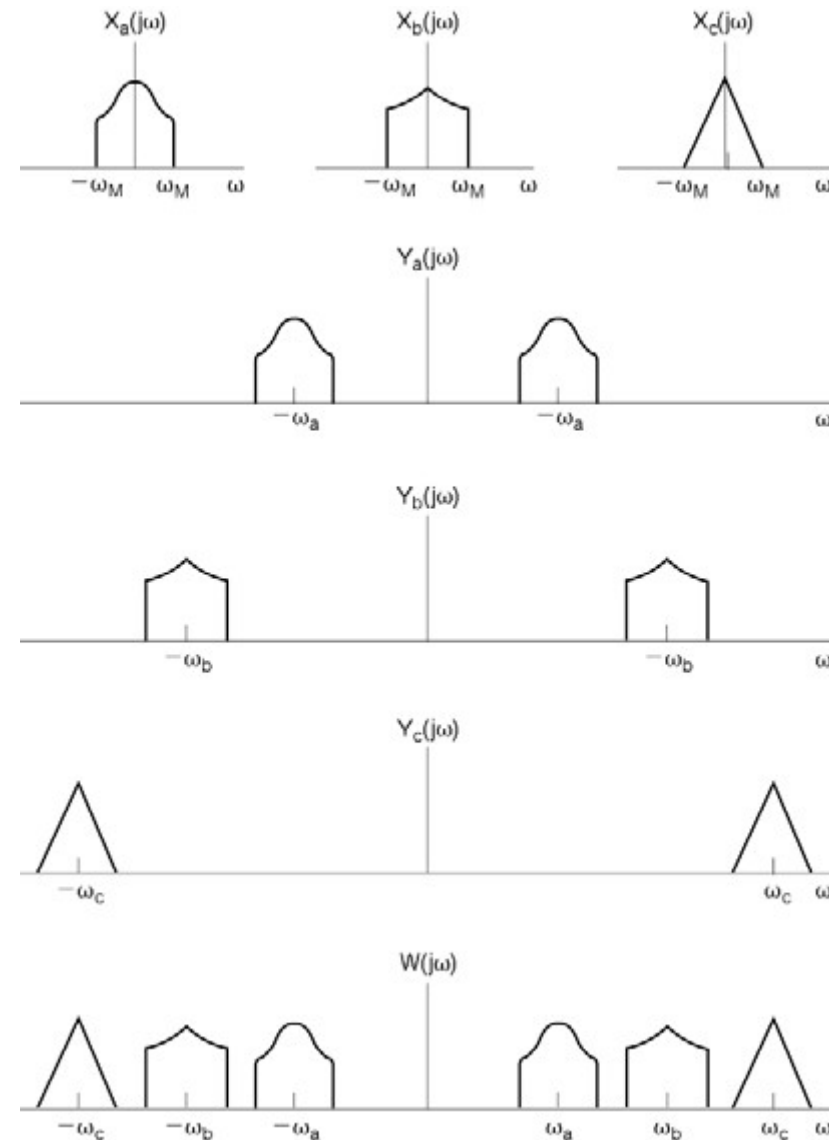
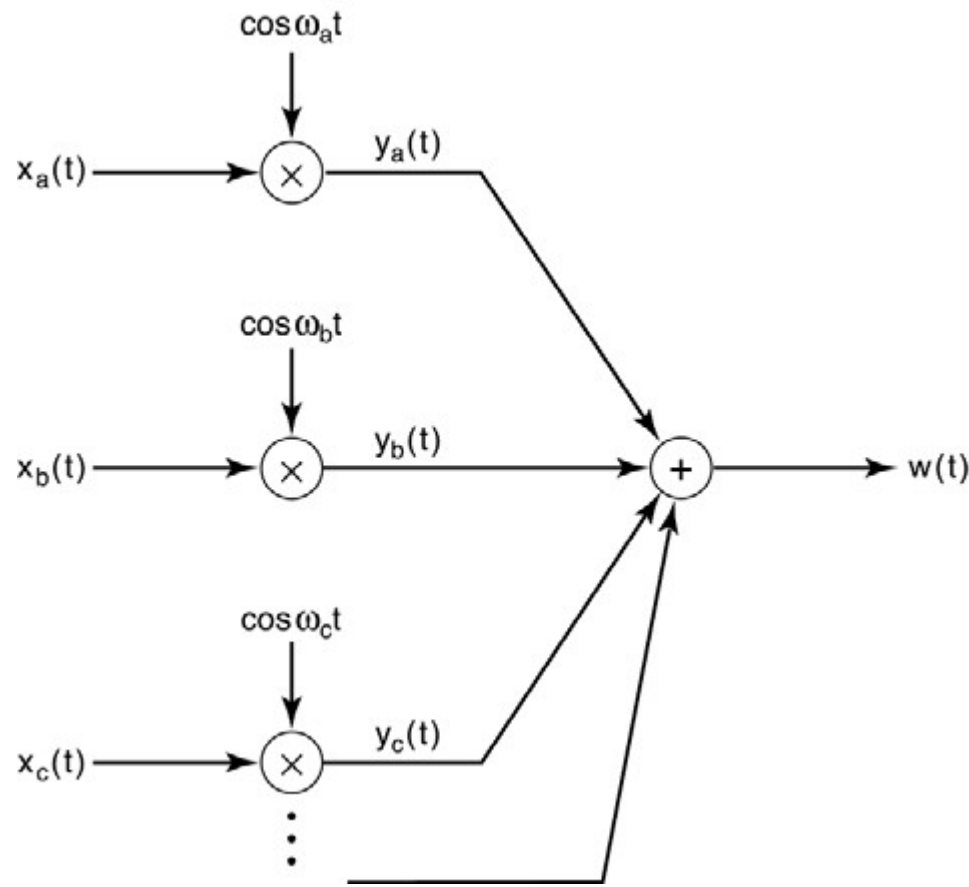


# Single-Sideband Modulation



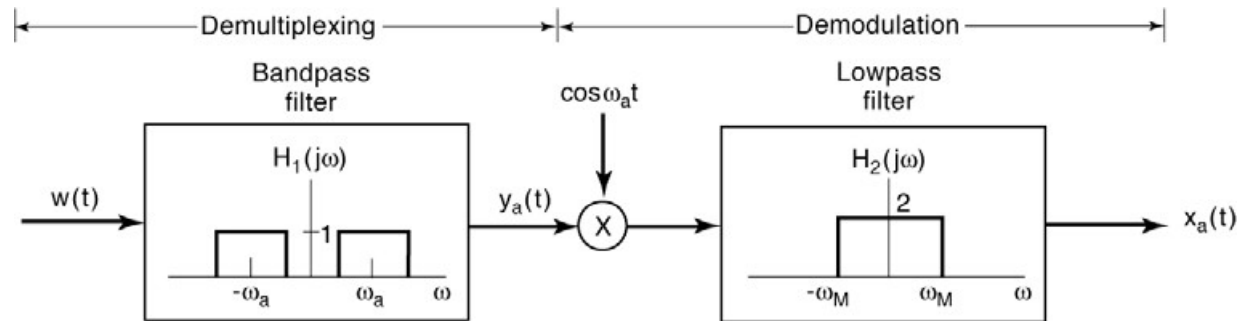
# Frequency Division Multiplexing

- Used in many communications systems including broadcast radio and cell phones.



# Demultiplexing and Demodulation

- Recall to recover one channel (e.g., radio station) from a multiplexed signal, we must first bandpass filter the multiplexed signal, and then demodulate it:



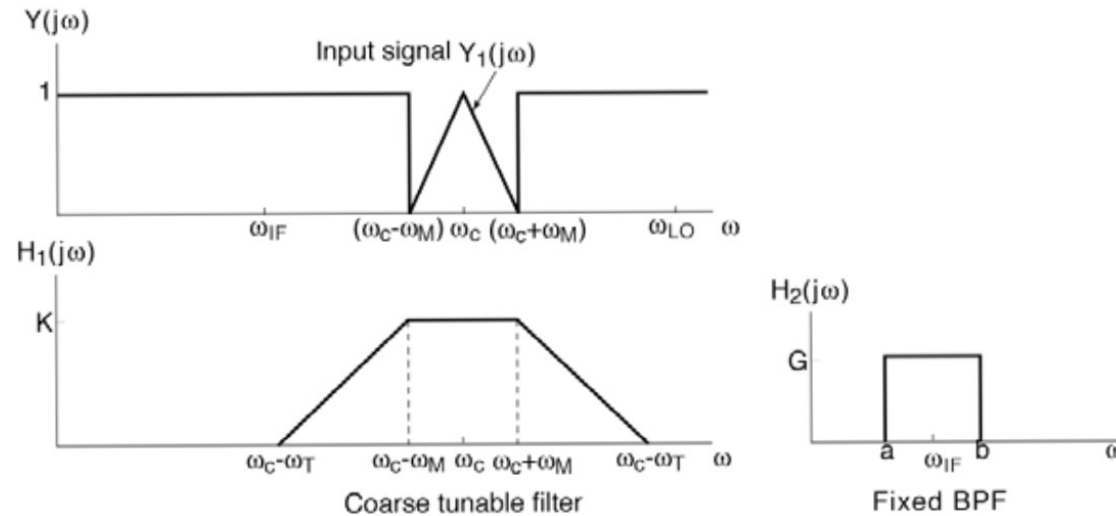
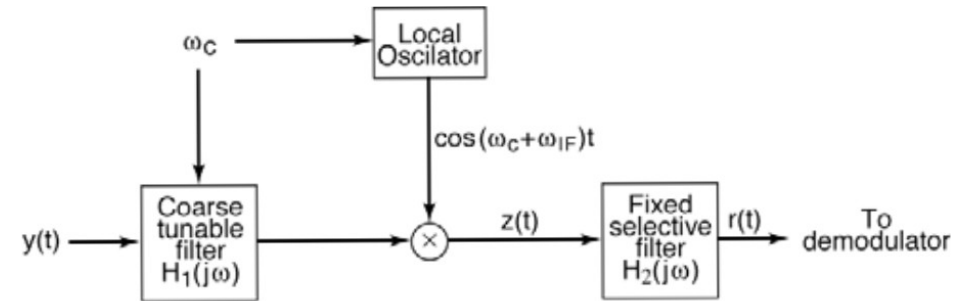
- However, the channels must not overlap for this to work. Fortunately, this is an important role played by the FCC (in the U.S.) – management of the electromagnetic spectrum.
- It is difficult to design a highly selective bandpass filter with a tunable center frequency.
- A better solution is the superheterodyne receiver: downconvert all channels to a common intermediate frequency. “The advantage to this method is that most of the radio's signal path has to be sensitive to only a narrow range of frequencies. Only the front end (the part before the frequency converter stage) needs to be sensitive to a wide frequency range.” ([Wiki](#))

# The Superheterodyne Receiver

AM Band: 535 – 1605 kHz

FCC-mandated

IF Frequency: 455 kHz



- **Principle: Down convert the received signal from  $\omega_c$  to  $\omega_{IF}$  using a coarse tunable bandpass filter. Use a sharp, fixed bandpass filter at  $\omega_{IF}$  to demodulate the remaining signal and remove remnants of the other channels that pass through the initial coarse filter.**