

# **20CYS301**Digital Communications L07-L09 Jul 23, 2024

Amplitude Modulation [AM]

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- I. DSB-SC AM
- II. DSB-SC AM Demodulation
- III. Conventional AM
- IV. Conventional AM Demodulation
- V. Quadrature Carrier Multiplexing

#### **DSB-SC AM [1/7]**

Amplitude Modulation is modulation of an analog carrier using analog data. Examples of such data include speech, music, images or videos.

#### We discuss:

- 1. how amplitude modulation can be constructed <u>mathematically</u> (the analog message signals change the amplitude of an analog carrier)
- 2. methods for <u>demodulation</u> of the carrier-modulated signal to recover the analog information

Treatment of system performance under various <u>noise conditions & power efficiency</u> will be considered after we discuss angle modulation.

m(t) is the analog signal to be transmitted. It is assumed to be a lowpass signal with bandwidth W. Thus, if |f| > W, M(f) = 0.

Power of the signal is 
$$P_m = \lim_{T \to \infty} \frac{1}{T} \int_{-T/2}^{T/2} |m(t)|^2 dt$$

Carrier is  $c(t) = A_c cos(2\pi f_c t + \varphi_c)$ 

Modulation changes m(t) from lowpass to bandpass, in the neighborhood of carrier frequency.

#### **DSB-SC AM [2/7]**

#### Types of AMs:

- 1. Double-sideband-suppressed-carrier DSB-SC AM
- 2. Conventional double-sideband AM
- 3. Single-sideband SSB AM
- 4. Vestigial-sideband VSB AM

DSB-SC AM: message signal is m(t).  $c(t) = m(t)A_c cos(2\pi f_c t + \varphi_c)$  is carrier signal. Ignore Tx-Rx synchronization issues. Thus,  $\varphi_c$  can be set to 0. Thus, Tx signal is:  $u(t) = A_c m(t) cos(2\pi f_c t)$ .

Note: multiplication in time-domain is convolution in frequency domain.

#### **Frequency Spectrum of DSB-SC AM Waveform:**

**Problem:** Prove that 
$$\Im[u(t)] = \frac{A_c}{2}[M(f - f_c) + M(f + f_c)]$$

Solution: 
$$\Im[u(t)] = \Im[A_c m(t) cos(2\pi f_c t)] = \frac{A_c}{2} \Im[m(t) \left(e^{j2\pi f_c t} + e^{-j2\pi f_c t}\right)\right]$$

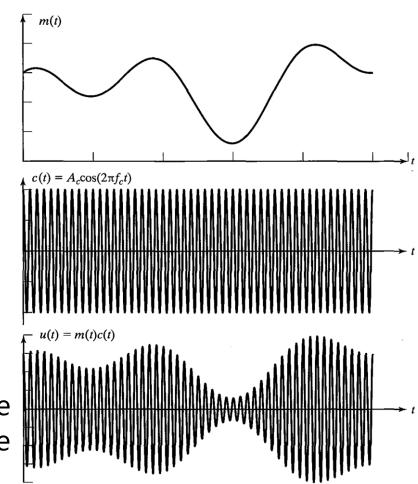
$$= \frac{A_c}{2} \Im \left[ m(t) e^{j2\pi f_c t} \right] + \frac{A_c}{2} \Im \left[ m(t) e^{-j2\pi f_c t} \right]$$

#### **DSB-SC AM [3/7]**

$$\begin{split} &= \frac{A_c}{2} \Im \left[ m(t) e^{j2\pi f_c t} \right] + \frac{A_c}{2} \Im \left[ m(t) e^{-j2\pi f_c t} \right] \\ &= \frac{A_c}{2} \frac{1}{T} \int_0^T m(t) e^{j2\pi f_c t} e^{-j2\pi f t} dt + \frac{A_c}{2} \frac{1}{T} \int_0^T m(t) e^{-j2\pi f_c t} e^{-j2\pi f t} dt \\ &= \frac{A_c}{2} \frac{1}{T} \int_0^T m(t) e^{j2\pi f_c t} e^{-j2\pi f t} dt + \frac{A_c}{2} \frac{1}{T} \int_0^T m(t) e^{-j2\pi f_c t} e^{-j2\pi f t} dt \\ &= \frac{A_c}{2} \frac{1}{T} \int_0^T m(t) e^{-j2\pi (f - f_c) t} dt + \frac{A_c}{2} \frac{1}{T} \int_0^T m(t) e^{-j2\pi (f + f_c) t} dt \end{split}$$

$$= \frac{A_c}{2}M(f - f_c) + \frac{A_c}{2}M(f + f_c) = \frac{A_c}{2}[M(f - f_c) + M(f + f_c)]$$

The spectrum of the message signal is shifted by  $f_c$ . If the message signal width was W then it occupies 2W after amplitude modulation (AM). Thus, channel bandwidth needed is  $B_c = 2W$ .

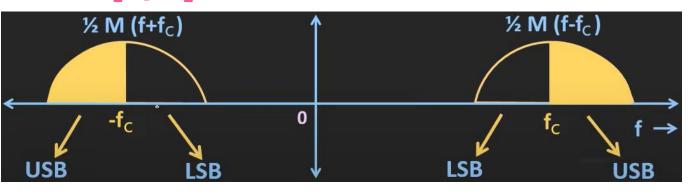


The frequency content of the modulated signal u(t) in the frequency band  $|f| > f_c$  is upper sideband of U(f) and that in  $|f| < f_c$  is lower sideband of U(f). As U(f) comprises both USB and LSB, it is called a double-sideband (DSB) AM signal.

#### **DSB-SC AM [4/7]**

$$U_f = \frac{A_c}{2} [M(f - f_c) + M(f + f_c)]$$
 upper side band (USB) & lower side band (LSB) of  $U_f$ .

Either of the two bands have all frequencies in M(f). Message is shifted by  $f_c$  and channel bandwidth is  $B_c = 2W$ .



Note, the carrier is not transmitted. Thus, that bit of power is conserved.

**Problem:** How do we know from  $U_f$  or the chart that the carrier was not transmitted?

**Solution:** Had the carrier been transmitted, you will see  $\delta(f \pm f_c)$  terms also in  $U_f$ .

Thus, the transmission is double side band (DSB) suppressed carrier (SC) amplitude modulated (AM) signal or, DSB SC AM signal, or, simply, DSBSC signal.

**Problem:** Express the product of 2 signals  $u_t = A\cos(2\pi f_m t)\cos(2\pi f_c t)$  as sum of two signals.

**Solution:** As  $A \cdot C[2\pi(f_m + f_c)t] = A \cdot C(2\pi f_m t)C(2\pi f_c t) - A \cdot S(2\pi f_m t)S(2\pi f_c t) \& A \cdot C[2\pi(f_m - f_c)t] = A \cdot C(2\pi f_m t)C(2\pi f_c t) + A \cdot S(2\pi f_m t)S(2\pi f_c t),$ 

$$\frac{1}{2}A \cdot C[2\pi(f_m + f_c)t] + \frac{1}{2}A \cdot C[2\pi(f_m - f_c)t] = A \cdot \cos(2\pi f_m t)\cos(2\pi f_c t)$$

#### **DSB-SC AM** [5/7]

**Problem:** In the previous problem, what if the signal is  $A \cdot cos(2\pi f_m t + \theta)cos(2\pi f_c t + \varphi)$ ?

**Solution:** DIY!

**Problem:** Find fourier transform of  $u_t = A \cdot cos(2\pi f_c t)$ .

Solution: 
$$\mathfrak{J}[u_t] = \mathfrak{J}[A \cdot C(2\pi f_c t)] = \frac{1}{2\pi} \int_{-\infty}^{\infty} A \cdot C(2\pi f_c t) e^{-j2\pi f_m t} dt$$

$$= \frac{1}{2\pi} \int_{-\infty}^{\infty} A \cdot \frac{C(2\pi f_c t) + jS(2\pi f_c t) + C(2\pi f_c t) - jS(2\pi f_c t)}{2} e^{-j2\pi f_m t} dt$$

$$= \frac{1}{2\pi} \int_{-\infty}^{\infty} A \cdot \left[ \frac{C(2\pi f_c t) + jS(2\pi f_c t)}{2} + \frac{C(2\pi f_c t) - jS(2\pi f_c t)}{2} \right] e^{-j2\pi f_m t} dt$$

$$= \frac{1}{2\pi} \int_{-\infty}^{\infty} A \cdot \left[ \frac{e^{j(2\pi f_c)t} + e^{-j(2\pi f_c)t}}{2} \right] e^{-j2\pi f_m t} dt = \frac{A}{4\pi} \int_{-\infty}^{\infty} e^{j2\pi (f_c - f_m)t} + e^{-j2\pi (f_c + f_m)t} dt$$

$$= \frac{A}{2} \left[ \frac{1}{2\pi} \int_{-\infty}^{\infty} e^{-j2\pi (f_m - f_c)t} dt \right] + \frac{A}{2} \left[ \frac{1}{2\pi} \int_{-\infty}^{\infty} e^{-j2\pi (f_c + f_m)t} dt \right]$$

$$= \frac{A}{2} \left[ \frac{1}{2\pi} \int_{-\infty}^{\infty} 1 \cdot e^{-j2\pi (f_m - f_c)t} dt \right] + \frac{A}{2} \left[ \frac{1}{2\pi} \int_{-\infty}^{\infty} 1 \cdot e^{-j2\pi (f_c + f_m)t} dt \right]$$

$$= \frac{A}{2}\delta(f - (f_m - f_c)) + \frac{A}{2}\delta(f - (f_m + f_c))$$

Thus, the Fourier Transformation of  $A \cdot cos(2\pi f_c t)$  are two delta functions of half the amplitude.

#### **DSB-SC AM [6/7]**

**Problem:** Message signal is  $m(t) = sinc(10^4 t)$ . Find DSB-SC-AM signal and its bandwidth if the carrier is a sinusoid of frequency 1 MHz, i.e.,  $c(t) = cos(2\pi 10^6 t)$ .

**Solution:** u(t) = m(t)c(t) If we can get the Fourier transform of m(t) then, as we know the signal is DSB-SC-AM signal, bandwidth of the signal will be 2 times the bandwidth of m(t).

$$\therefore \, \Im[m(t)] = \Im[sinc(10^4 t)] = \frac{1}{|10^4|} \, \Pi\left[\frac{f}{10^4}\right] \, \because \, \Im[at] = \frac{1}{|a|} \, F\left[\frac{f}{a}\right] \, \text{and} \, F[f] \, \text{is rectangle function} \, \Pi.$$

The rectangle function has a bandwidth of  $10^4$  Hz, centered at 0. Thus, DSB-SC-AM signal bandwidth is 10KHz.

What about "Find DSB-SC-AM signal"?

**Problem:** What is the power of DSB-SC-AM signal?

Solution: 
$$P[u(t)] = \lim_{T \to \infty} \frac{1}{T} \int_{-T/2}^{T/2} u^{2}(t) dt = \lim_{T \to \infty} \frac{1}{T} \int_{-T/2}^{T/2} A_{c}^{2} m^{2}(t) C^{2}(2\pi f_{c}t) dt$$

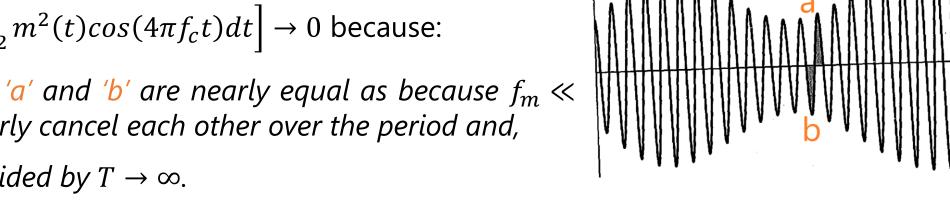
$$= \lim_{T \to \infty} \frac{1}{T} \int_{-T/2}^{T/2} \frac{A_{c}^{2}}{2} m^{2}(t) [1 + \cos(4\pi f_{c}t)] dt \dots \therefore C^{2}(t) = \frac{[1 + C(2T)]}{2}$$

$$= \frac{A_{c}^{2}}{2} \left[ \lim_{T \to \infty} \frac{1}{T} \int_{-T/2}^{T/2} m^{2}(t) dt \right] + \frac{A_{c}^{2}}{2} \left[ \lim_{T \to \infty} \frac{1}{T} \int_{-T/2}^{T/2} m^{2}(t) \cos(4\pi f_{c}t) dt \right] \approx \frac{A_{c}^{2}}{2} P_{m}$$

# **DSB-SC AM [7/7]**

$$\frac{A_c^2}{2} \left[ \lim_{T \to \infty} \frac{1}{T} \int_{-T/2}^{T/2} m^2(t) \cos(4\pi f_c t) dt \right] \to 0 \text{ because:}$$

- a. The two areas 'a' and 'b' are nearly equal as because  $f_m \ll$  $f_c$ , the two nearly cancel each other over the period and,
- **b**. The term is divided by  $T \to \infty$ .



**Problem:** What is the power of USB-SC-AM and LSB-SC-AM signal?

**Solution:** Do it yourself!

Plot of  $m^2(t) \cos(4\pi f_c t)$ 

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### **DSB-SC AM Demodulation [1/2]**

If a DSB-SC AM signal u(t) is transmitted through an ideal channel then the received signal,  $r(t) = u(t) = A_c m(t) \cos(2\pi f_c t).$ 

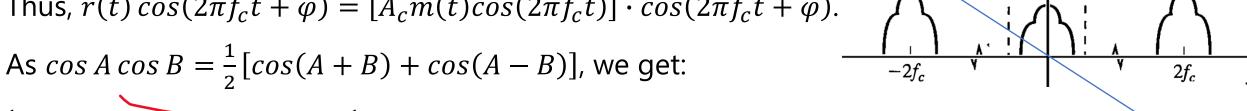
**Ideal channel:** it may be defined as an impulse function with a sampling frequency at least twice of the maximum frequency in m(t).

To demodulate, we multiply r(t) with a locally generated sinusoid.

Assume the receiver crystal operates at the same frequency as that of the carrier,  $f_c$ , but is out of sync in phase with the transmitter by  $\varphi$ .

Thus, 
$$r(t) cos(2\pi f_c t + \varphi) = [A_c m(t) cos(2\pi f_c t)] \cdot cos(2\pi f_c t + \varphi)$$
.

As 
$$\cos A \cos B = \frac{1}{2} [\cos(A+B) + \cos(A-B)]$$
, we get:



$$\frac{1}{2}A_c m(t)\cos(4\pi f_c t + \varphi) + \frac{1}{2}A_c m(t)\cos(\varphi)$$
 ... the 1<sup>st</sup> term can be filtered out by a low pass filter.

This is possible in a cost-effective manner only because signal bandwidth  $W \ll f_c$ .

The output of an ideal DSB-SC AM lowpass filter is  $\frac{1}{2}A_c m(t) \cos(\varphi)$ .

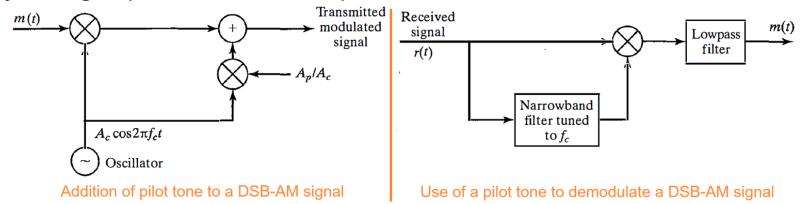
### **DSB-SC AM Demodulation [2/2]**

Note that the demodulated signal is weaker. Its power decreases by a factor of  $cos^2(\varphi)$ .

If Rx is 45° out of phase with the Tx then, the power of demodulated signal is  $cos^2(45^\circ) = (1/\sqrt{2})^2 = 50\%$  of the received signal.

This can be addressed in two ways:

- 1. Adding a carrier component, called a pilot tone, to the carrier component whose magnitude,  $A_p$  (power is  $A_p^2/2$ ), is much smaller than that of the modulated signal. This introduces a DC component and the signal is no longer a DSB-SC AM signal.
- 2. Generating a phase-locked sinusoidal carrier from the received signal r(t) without the need of a pilot signal by using a phase-locked loop (TBD later).



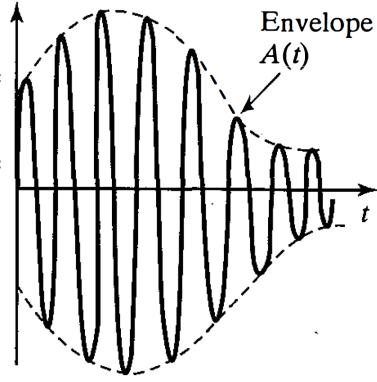
- I. DSB-SC AM
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Conventional AM [1/4] Recall that the received signal for DSB-SC AM was  $r(t) = u(t) = A_c m(t) \cos(2\pi f_c t)$ .  $A_c m(t) \cos(2\pi f_c t)$ .

In case of conventional AM, i.e., just AM, it is r(t) = u(t) = $A_c[1 + m(t)] \cos(2\pi f_c t)$  where  $|m(t)| \le 1$ .

If  $|m(t)| \le 1$ , then  $A_c[1+m(t)]$  is +ve & demodulation is easier.  $\therefore$  commercial broadcasters use this kind of modulation.

If |m(t)| > 1, then  $A_c[1 + m(t)]$  is -ve. This signal is called overmodulated & its demodulation is complex.



$$m(t) = am_n(t); -1 \le m_n(t) \le 1 \text{ and } 0 \le a \le 1. \text{ Thus, } r(t) = u(t) = A_c[1 + am_n(t)] \cos(2\pi f_c t).$$

**Spectrum of AM Signal:** 
$$\Im[u(t)] = U(f) = \Im[A_c[1 + am_n(t)] \cos(2\pi f_c t)]$$

$$= \Im[A_c a m_n(t) \cos(2\pi f_c t)] + \Im[A_c \cos(2\pi f_c t)]$$

$$= \frac{A_c a}{2} [M_n(f - f_c) + M_n(f + f_c)] + \frac{A_c}{2} [\delta(f - f_c) + \delta(f + f_c)]$$

Note: Conventional AM is same as DSB-SC where m(t) is replaced by  $[1 + am_n(t)]$ 

# **Conventional AM [2/4]**

Like in the case of DSB-SC AM, conventional Message signal m(t).

AM will occupy twice the bandwidth of the message signal m(t).

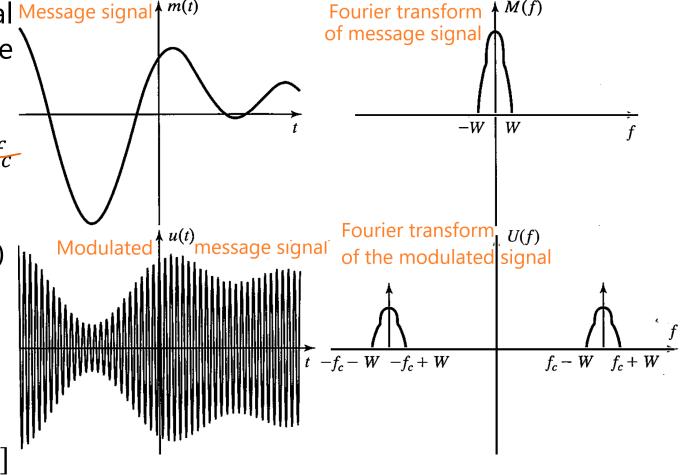
If message signal  $m(t) = cos(2\pi f_m t)$ ;  $f_m \ll f_c$ then,  $u(t) = A_c[1 + a cos(2\pi f_m t)] cos(2\pi f_c t)$ 

$$= A_c cos(2\pi f_c t) + A_c a cos(2\pi f_m t) cos(2\pi f_c t) \parallel$$

$$u(t) = A_c \cos(2\pi f_c t) + \frac{A_c a}{2} \cos[2\pi (f_c - f_m)t] + \frac{A_c a}{2} \cos[2\pi (f_c + f_m)t]$$

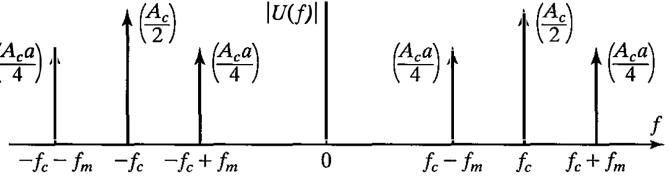
LSB component:  $u_l(t) = \frac{A_c a}{2} cos[2\pi (f_c - f_m)t]$ 

USB component:  $u_l(t) = \frac{A_c a}{2} cos[2\pi (f_c + f_m)t]$ 



#### **Conventional AM [3/4]**

**Problem:** Compute power of conventional AM signal using u(t) derived on last slide. Is it the same if U(f) is used? **Solution:** DIY. The answer is  $A_c^2/2 + a^2A_c^2/4$ . Note that carrier power  $\geq 2x$  of message signal.  $-f_c - f_m - f_c - f_c + f_m$ 



Depicted in the figure is the spectrum of conventional AM signal, i.e., DSB AM signal.

Recall that received power of DSB-SC AM is  $\frac{A_c^2}{2}P_m$ . Compare this with received power of DSB AM of  $\frac{A_c^2}{2} + \frac{a^2A_c^2}{4}$  if the message signal is  $cos(2\pi f_m t)$  or,  $\frac{A_c^2}{2} + \frac{a^2A_c^2}{2}P_m$  for a general case.

$$\frac{Efficiency_{DSB-AM}}{Efficiency_{DSB-SCAM}} = \frac{\frac{A_c^2}{2} + \frac{a^2A_c^2}{2}P_m}{\frac{A_c^2}{2}P_m} = \frac{1}{P_m} + a^2 \dots \text{ why did we cancel out the terms?}$$

As a is typically quite small, power efficiency of conventional demodulators is much smaller than that of DSB-SC AM transmissions. Still, the former is preferred due to ease in demodulation, which in this case, makes receiver demodulators materially cheaper.

# **Conventional AM [4/4]**

**Problem:** Message signal  $m(t) = 3 \cos(200\pi t) + \sin(600\pi t)$ .

Carrier signal  $c(t) = cos(2 \times 10^5 t)$ . Modulation index a = 0.85.



**Solution:** Step 1: find extremum of the signal. To do this, take derivative and set it to 0.

Thus, 
$$\frac{d(m(t))}{dt} = 3 \cdot 200\pi \left[ -\sin(200\pi t) \right] + 600\pi \cdot \cos(600\pi t) = 0$$
.  $\therefore \cos(600\pi t) = \sin(200\pi t)$ .

As 
$$sin(\theta) = cos(\frac{\pi}{2} - \theta)$$
,  $cos(600\pi t) = cos(\frac{\pi}{2} - 200\pi t)$  :  $cos^{-1}[cos(600\pi t)] = cos^{-1}[cos(\frac{\pi}{2} - 200\pi t)]$ 

$$[200\pi t]$$
, i.e.,  $600\pi t = \frac{\pi}{2} - 200\pi t$ , i.e.,  $t = \frac{1}{1600}$ .

$$m(t)_{|t=\frac{1}{1600}} = 3\cos\left(200\pi \cdot \frac{1}{1600}\right) + \sin\left(600\pi \cdot \frac{1}{1600}\right) = 3.696$$

Thus, maximum value of normalized message signal is  $m_n(t) = \frac{3\cos(200\pi t) + \sin(600\pi t)}{3.696} = 0.82\cos(200\pi t) + 0.27\sin(600\pi t)$ .

Power in sum of 2 sinusoids is the sum of powers of each sinusoid, i.e.,  $P_{m_n} = \frac{1}{2}[0.82^2 + 1.2]$ 

$$[0.27^2] = 0.37$$
. Carrier power:  $\frac{A_c^2}{2} = 0.5$  & sideband power:  $\frac{A_c^2 \cdot a^2 \cdot P_{mn}}{2} = \frac{1}{2} \times 0.85^2 \times 0.37 = 0.133$ .

# **Conventional AM [explanatory note for last slide]**

In these slides on slide "Conventional AM [4/4]", the numerical is indeed correct.

The integration of the cross term, 0.44  $cos(200\pi t)*sin(600\pi t)$  indeed goes to zero when you integrate it.

.: the statement "Power in sum of 2 sinusoids is the sum of powers of each sinusoid" is correct.

One can convince oneself of this using 3 methods:

- do it via actual integration or
- graphically assess it as on slide "DSB-SC AM [7/7]" or,
- by looking at the figure on slide "Conventional AM [3/4]"

In the exam, you do not need to show how the integration of the cross terms go to zero.

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### **Conventional AM Demodulation [1/1]**

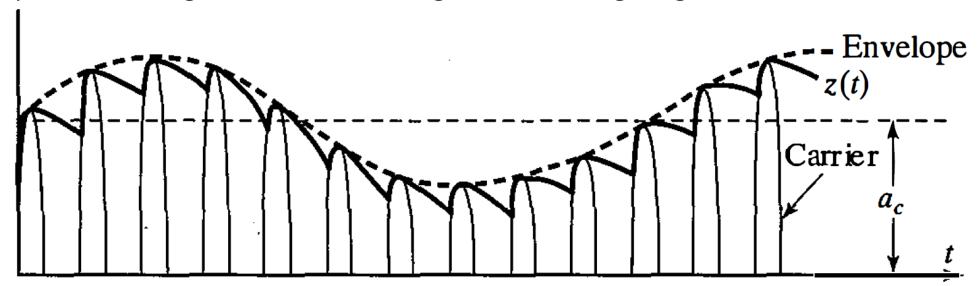
**Demodulation of Conventional DSB-AM Signal:** The demodulation is easier than that of DSB-SC AM as there is no need to synchronize demodulator.

As long as |m(t)| < 1, the envelope 1 + m(t) > 0.

If we rectify the signal then the negative values are eliminated.

The rectified signal is then passed through a low-pass filter whose bandwidth matches that of the message signal.

Rectifier followed by lowpass filter is called *envelop detector*. The envelope then must be passed through a DC blocker to get the message signal.



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# **Quadrature Carrier Multiplexing [1/3]**

**Signal Multiplexing:** If we wish to send  $\geq 2$  messages simultaneously then you could modulate each with a different carrier frequency.

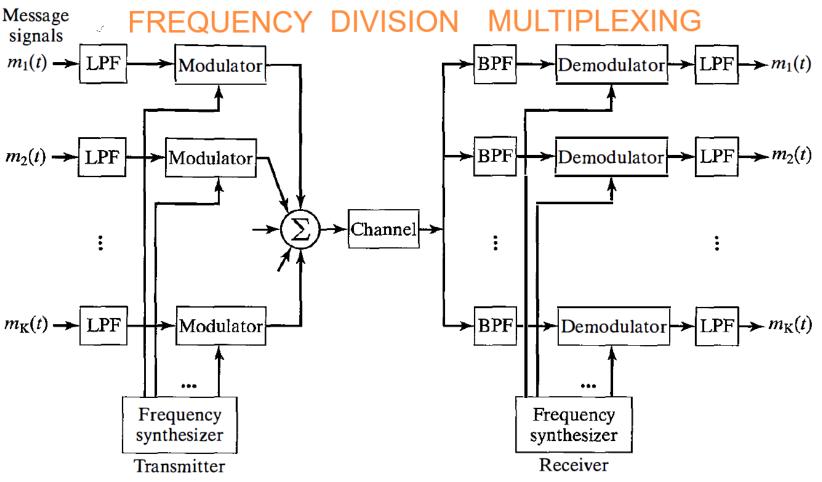
**Problem:** Is this an example of time division or frequency division multiplexing? Justify.

**Solution:** DIY.

**Problem:** What is the min distance between carrier frequencies for SSB? for DSB? Justify.

**Solution:** DIY.

Time division multiplexing is typically not used for transmitting analog information.



# **Quadrature Carrier Multiplexing [2/3]**

Quadrature Carrier Multiplexing (QCM): We can instead, send the data on two carriers:  $A_c cos(2\pi f_c t)$  and  $A_c sin(2\pi f_c t)$ . Thus,  $u(t) = A_c m_1(t) cos(2\pi f_c t) + A_c m_2(t) sin(2\pi f_c t)$ .

**Problem:** Bandwidth efficiency of QCM is higher or lower or the same as that of SSB?

**Solution: DIY** 

A Balanced Modulator is a (de)modulator that implements DSB SC modulation.

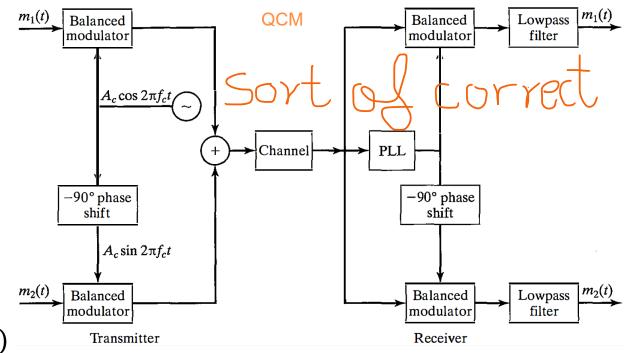
PLL is phase-locked loop.

**Demodulation:** multiply by  $cos(2\pi f_c t)$ .

$$\therefore u(t)\cos(2\pi f_c t) = A_c m_1(t)\cos^2(2\pi f_c t) + A_c m_2(t)\sin(2\pi f_c t)\cos(2\pi f_c t)$$

$$= \frac{A_c}{2} m_1(t) \cdot [1 + \cos(4\pi f_c t)] + \frac{A_c m_2(t) \sin(4\pi f_c t)}{2}$$

$$= \frac{A_c}{2}m_1(t) + \frac{A_c}{2}m_1(t)\cos(4\pi f_c t) + \frac{A_c}{2}m_2(t)\sin(4\pi f_c t)$$



# **Quadrature Carrier Multiplexing [3/3]**

Low pass filter will remove the highlighted term, leaving only the envelope of  $m_1(t)$ .

**Problem:** How do we recover  $m_2(t)$ ?

**Solution:** Multiply  $u(t) = A_c m_1(t) cos(2\pi f_c t) + A_c m_2(t) sin(2\pi f_c t)$  with  $sin(2\pi f_c t)$ .

Thus, 
$$u(t)cos(2\pi f_c t) = A_c m_1(t)cos(2\pi f_c t)sin(2\pi f_c t) + A_c m_2(t)sin^2(2\pi f_c t)$$

$$= \frac{A_c}{2}m_1(t)\sin(4\pi f_c t) + A_c m_2(t)\sin^2(2\pi f_c t)$$

$$= \frac{A_c}{2}m_1(t)\sin(4\pi f_c t) + A_c m_2(t)[1 - \cos^2(2\pi f_c t)]$$

$$= \frac{A_c}{2}m_1(t)\sin(4\pi f_c t) + A_c m_2(t)\left[1 - \frac{1 + \cos(4\pi f_c t)}{2}\right] = \frac{A_c}{2}m_1(t)\sin(4\pi f_c t) + A_c m_2(t)\frac{1 - \cos(4\pi f_c t)}{2}$$

$$= \frac{A_c}{2} m_1(t) \sin(4\pi f_c t) + \frac{A_c}{2} m_2(t) - \frac{A_c}{2} m_2(t) \cos(4\pi f_c t)$$

A low pass filter eliminates the highlighted terms, leaving just the envelop of  $m_2$ .