

# Communication Theory

## English Grades

## Chapter 2

### Analog Modulations

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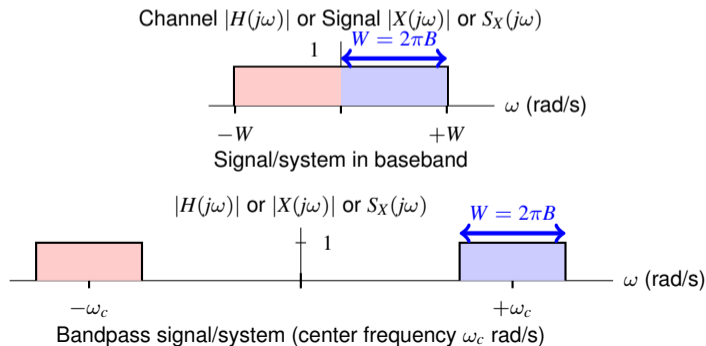
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## Bandwidth of a signal/system

- **Baseband** signal/system
  - ▶ Frequency response is centered at 0 Hz
- **Bandpass** signal/system
  - ▶ Frequency response is centered at center frequency  $f_c$  Hz
  - ▶ Or, equivalently, at  $\omega_c = 2\pi f_c$  rad/s
- **Bandwidth** ( $B$  Hz, or  $W = 2\pi B$  rad/s)
  - ▶ Range of **positive frequencies** that are used or available



## Analog signal to be transmitted: modulating signal $m(t)$

- Deterministic signal: characteristics of signal  $m(t)$ 
  - ▶ Low pass signal of **bandwidth  $B$  Hz** (or  $W = 2\pi B$  rad/s)
    - ★ Fourier transform  $M(j\omega)$  with  $M(j\omega) = 0$  for  $|\omega| > 2\pi B$
  - ▶ It is a power signal: Its power is  $P_m$  Watts

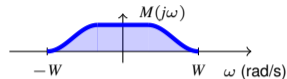
$$P_m = \lim_{T \rightarrow \infty} \frac{1}{T} \int_{-T/2}^{+T/2} |m(t)|^2 dt$$

- Random (stochastic) signal: statistical analysis (“average”)
  - ▶ Model for the signal: stochastic process  $M(t)$ 
    - ★ Wide Sense Stationary Process (WSS)
    - ★ Zero mean  $m_M = 0$
    - ★ Autocorrelation function  $R_M(\tau)$
    - ★ Power spectral density  $S_M(j\omega)$
    - ★ **Bandwidth  $B$  Hz** :  $S_M(j\omega) = 0$  for  $|\omega| > 2\pi B$
    - ★ Power:  $P_M$  Watts

$$P_M = R_M(0) = \frac{1}{2\pi} \int_{-\infty}^{\infty} S_M(j\omega) d\omega$$

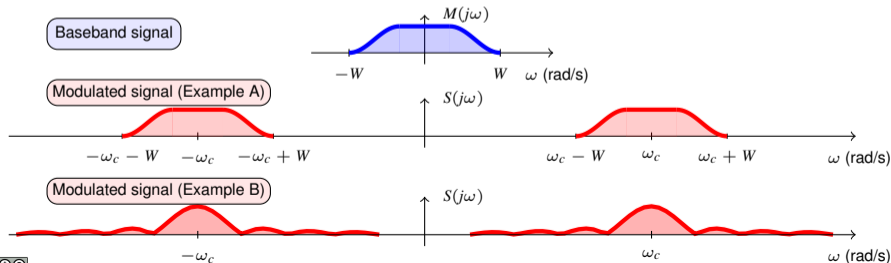
# Analog Communications Systems

- Information signal (modulating signal)  $m(t)$ 
  - ▶ Analog signal: the information is in the waveform  $m(t)$
  - ▶ Baseband and bandlimited: Bandwidth  $W$  rad/s



- Types of transmission in analog systems

- ▶ Baseband transmission (unmodulated)
- ▶ Modulated transmission :  $s(t)$ 
  - ★ Spectrum of the signal is shifted (center or carrier frequency  $\omega_c$ )
  - ★ The shape can be maintained or changed (usually bandwidth can be spread)

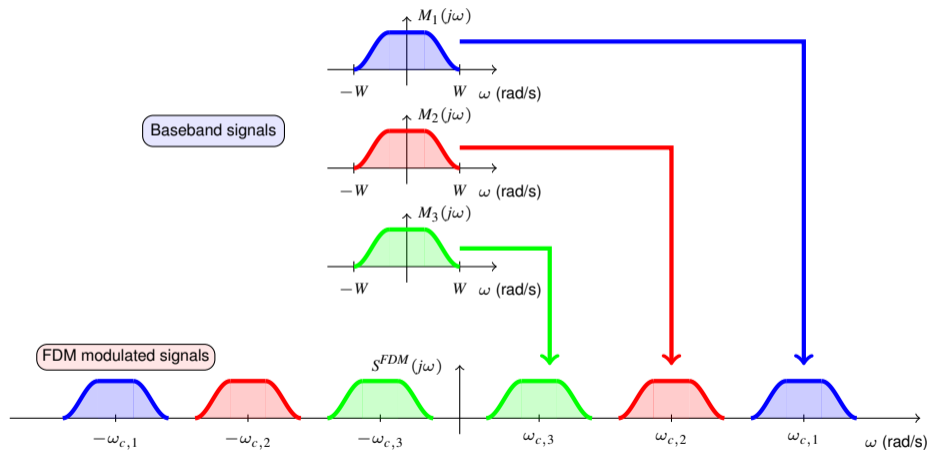


# Purpose of modulating an analog signal

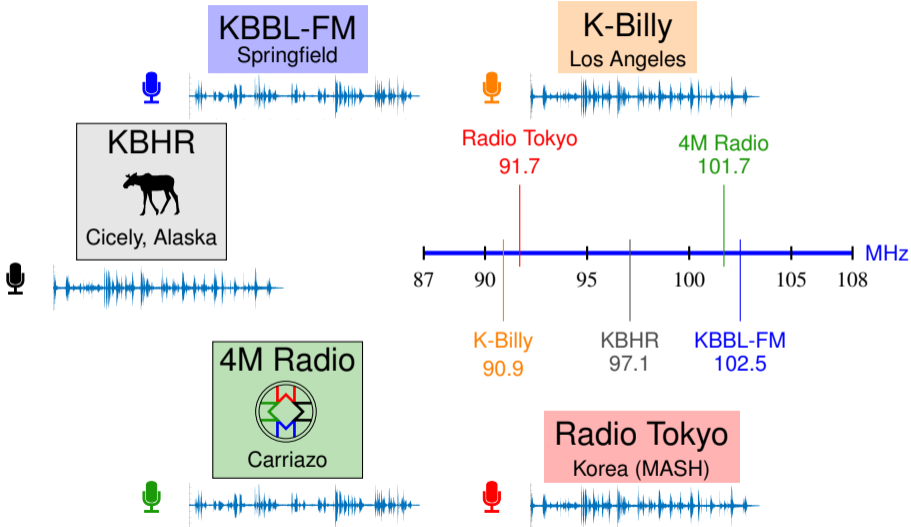
- 1 Adapting the signal to the characteristics of the channel
  - ▶ Choose the appropriate and/or available frequency range
    - ★ The spectrum of the signal is shifted to the corresponding band
- 2 Expanding the bandwidth
  - ▶ Increase the noise immunity of the modulated signal during transmission
- 3 Multiplexing / medium access
  - ▶ Accommodating the simultaneous transmission of different signals on the same medium
    - ★ Frequency Division Multiplexing (FDM)
    - ★ Frequency Division Multiple Access (FDMA)

# Frequency Division Multiplexing (FDM)

- The spectrum of different signals is shifted to non-overlapping frequency bands



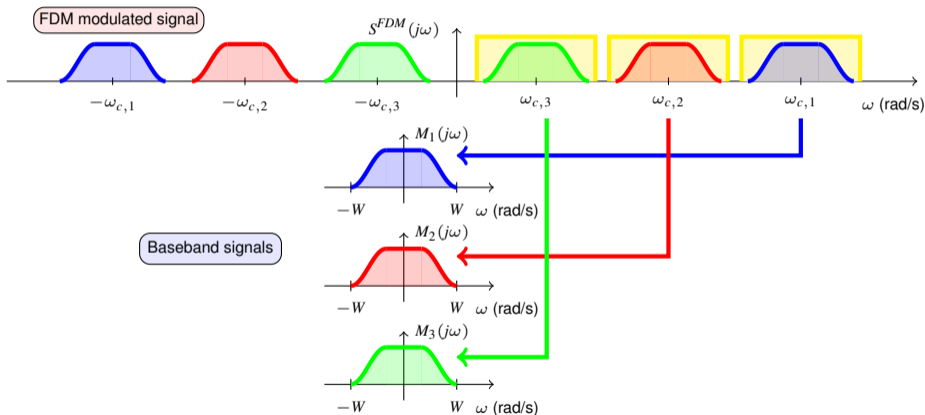
# FDM - Example: Commercial Radio





# FDM - Demultiplexing

- At the receiver, the spectrum of each signal is filtered and shifted back to baseband, allowing each signal to be recovered



# Introduction to the concept of modulation

## ● Analog modulation:

- ▶ The analog information (modulating) signal is printed in a sinusoidal carrier

$$c(t) = A_c \cos(\underbrace{2\pi f_c t + \phi_c}_{\omega_c}) \left\{ \begin{array}{l} \text{Amplitude : } A_c \\ \text{Phase : } \phi_c \\ \text{Frequency: } f_c \text{ or } \omega_c \end{array} \right.$$

- ▶ Parameters of a sinusoidal carrier:

- ★ Amplitude:  $A_c$  V
- ★ Phase:  $\phi_c$  rad
- ★ Frequency:  $f_c$  Hz (or equivalently,  $\omega_c = 2\pi f_c$  rad/s)

## ● Types of analog modulations

- ▶ Amplitude Modulation (AM)  $A_c \rightarrow A_c(t) = f(m(t))$
- ▶ Angle modulations
  - ★ Phase Modulation (PM)  $\phi_c \rightarrow \phi_c(t) = f(m(t))$
  - ★ Frequency Modulation (FM)  $f_i(t) = f_c \rightarrow f_i(t) = f(m(t))$   
 $f_i(t)$ : instantaneous frequency of the carrier signal

## Amplitude Modulations (AM)

- The modulating signal (or message)  $m(t)$  is encoded in the amplitude of the carrier signal  $c(t)$

$$c(t) = A_c \cos(\omega_c t + \phi_c)$$

$$A_c \rightarrow A_c(t) = f(m(t))$$

- There are different variants of AM modulation
  - ▶ AM: Conventional AM modulation (double sideband with carrier)
  - ▶ DSB: Double SideBand (without carrier)
  - ▶ SSB: Single SideBand
  - ▶ VSB: Vestigial SideBand

## Conventional AM modulation

- Double SideBand (DSB) + carrier

$$s(t) = \underbrace{A_c \cos(\omega_c t + \phi_c)}_{\text{Carrier } c(t)} + \underbrace{m(t) \times A_c \cos(\omega_c t + \phi_c)}_{\text{Double Sideband (DSB): } m(t) \times c(t)}$$

$$s(t) = A_c [1 + m(t)] \cos(\omega_c t + \phi_c)$$

- ▶ Desirable situation: envelope proportional to  $m(t)$

$$\text{If } A_c [1 + m(t)] \geq 0 \forall t \Rightarrow \text{Envelope} \equiv A_c [1 + m(t)]$$

- Overmodulation: occurs when  $A_c [1 + m(t)] < 0$  at some instants

- ▶ Cause:

- ★ Occurs when  $m(t) < -1$

- ▶ Effect:

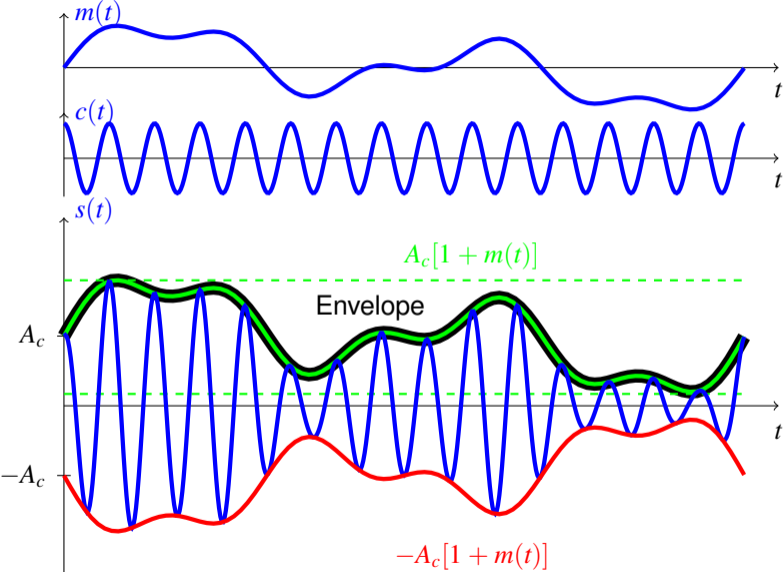
- ★ The carrier is inverted at those instants

- ★ The envelope becomes  $-A_c [1 + m(t)]$

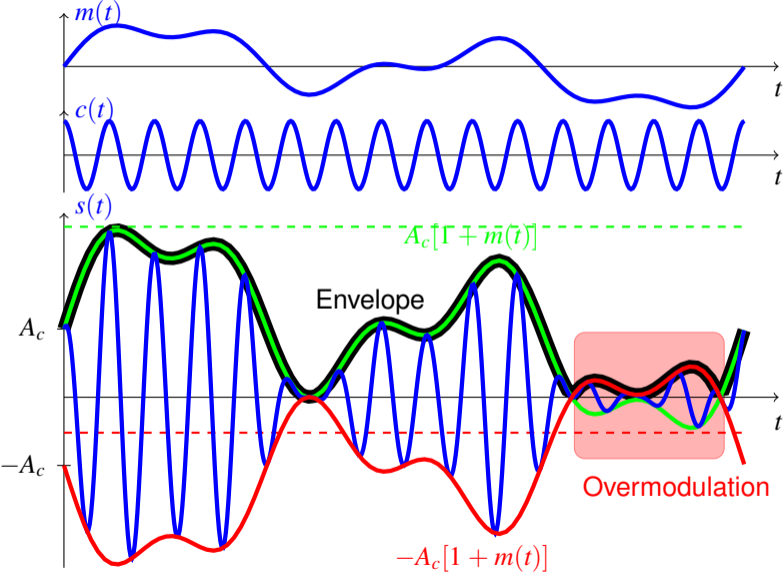
- ▶ Solution:

$$\text{Ensure that } |m(t)| \leq 1 \begin{cases} \text{Message normalization: } m_n(t) \\ \text{Introduction of the MODULATION INDEX: } a \end{cases}$$

# AM modulation without overmodulation



# AM modulation with overmodulation



## Conventional AM Modulation - Modulation Index

- Normalized modulating signal (message)  $m_n(t)$

$$m_n(t) = \frac{m(t)}{\max |m(t)|} = \frac{m(t)}{C_M}$$

- $C_M = \max |m(t)|$ : Range of  $m(t)$ :  $-C_M \leq m(t) \leq +C_M$
- Modulation index ( $a$ )
  - $m(t)$  is replaced by modulating signal with modulation index  $a$

$$m_a(t) = a \times m_n(t) = \frac{a}{C_M} \times m(t)$$

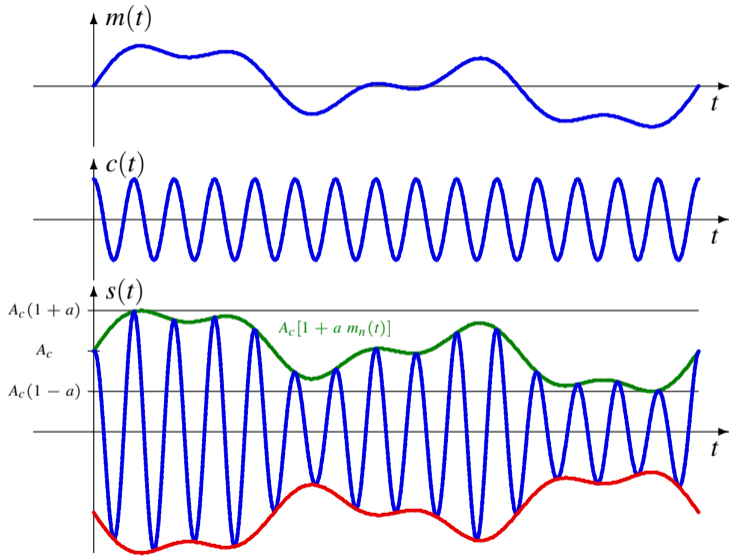
Range of  $m_a(t)$ :  $-a \leq m_a(t) \leq +a$

- ★ To avoid overmodulation:  $0 < a \leq 1$

- Modulated signal with modulation index  $a$

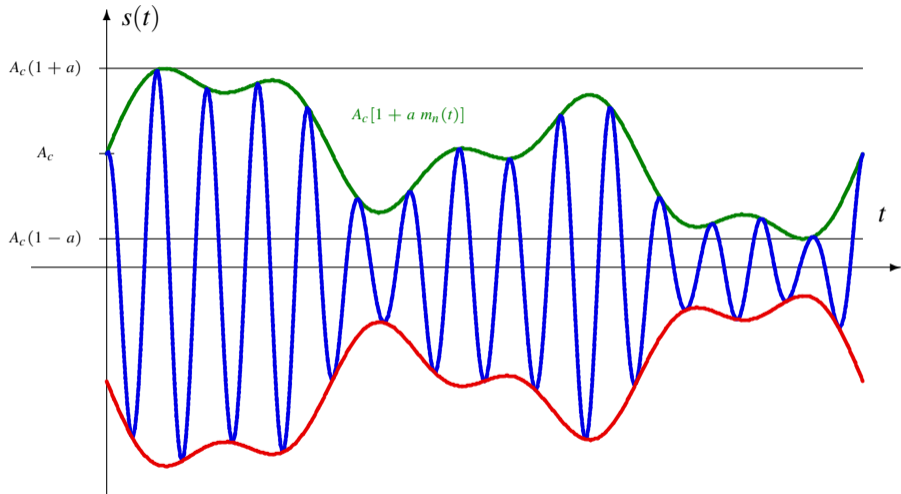
$$\begin{aligned} s(t) &= A_c [1 + \underbrace{a m_n(t)}_{m_a(t)}] \cos(\omega_c t + \phi_c) \\ &= A_c \cos(\omega_c t + \phi_c) + m_a(t) \times A_c \cos(\omega_c t + \phi_c) \end{aligned}$$

# Waveform of a conventional AM modulation

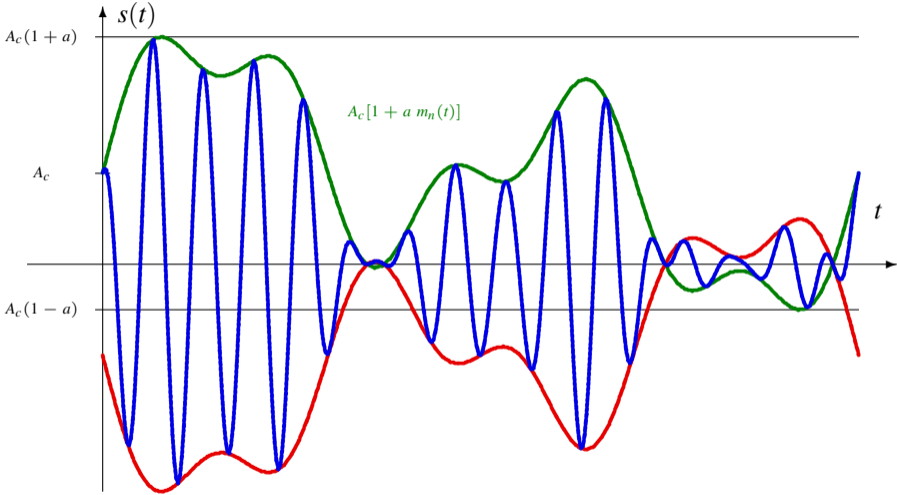




# Waveform of an AM modulation ( $a = 0.75$ )



# Overmodulation ( $a = 1.5$ )



## Spectrum of the AM signal - Deterministic

- Conventional AM signal:  $s(t) = c(t) + m_a(t) \times c(t)$

$$s(t) = A_c \cos(\omega_c t + \phi_c) + \underbrace{m_a(t)}_{\frac{a}{C_M} m(t)} \times A_c \cos(\omega_c t + \phi_c)$$

- Deterministic signal  $m(t)$  with FT  $M(j\omega)$ ,  $M(j\omega) = 0$  for  $|\omega| > 2\pi B$  rad/s

- Spectrum of  $m_a(t) = a m_n(t)$ :  $M_a(j\omega) = a M_n(j\omega) = \frac{a}{C_M} M(j\omega)$

- Conventional AM signal spectrum

$$\begin{aligned} S(j\omega) &= \mathcal{FT}\{A_c \cos(\omega_c t + \phi_c)\} + \frac{1}{2\pi} \mathcal{FT}\{m_a(t)\} * \mathcal{FT}\{A_c \cos(\omega_c t + \phi_c)\} \\ &= A_c \pi [\delta(\omega - \omega_c) e^{j\phi_c} + \delta(\omega + \omega_c) e^{-j\phi_c}] \\ &\quad + \frac{A_c}{2} \left[ \underbrace{M_a(j\omega - j\omega_c)}_{\frac{a}{C_M} M(j\omega - j\omega_c)} e^{j\phi_c} + \underbrace{M_a(j\omega + j\omega_c)}_{\frac{a}{C_M} M(j\omega + j\omega_c)} e^{-j\phi_c} \right] \end{aligned}$$

## Spectrum of the AM signal - Analysis

- Modulus of the Fourier transform  $S(j\omega)$ 
  - ▶ Two deltas, in  $-\omega_c$  and in  $+\omega_c$ 
    - ★ Amplitude  $A_c\pi$
  - ▶ Replicas of the form of  $M(j\omega)$  shifted to  $-\omega_c$  and  $+\omega_c$ 
    - ★ Scale factor  $\frac{A_c a}{2 C_M}$
- Phase of the Fourier transform
  - ▶ The carrier phase introduces the terms  $e^{\pm j\phi_c}$ 
    - ★ Linear phase terms
- Bandwidth of the modulated signal

$$W_{AM} = 2 W \text{ rad/s}$$

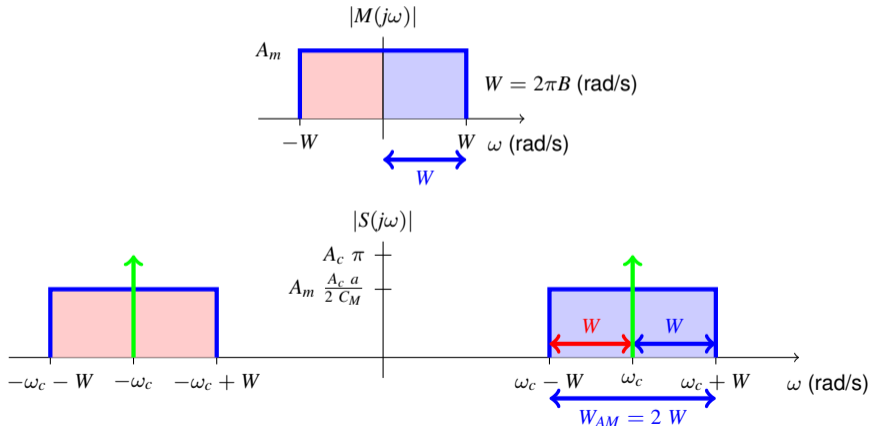
$$B_{AM} = 2 B \text{ Hz}$$

- ▶ Twice the bandwidth of the modulating signal  $m(t)$

# Spectrum of conventional AM signal - Representation

$$S(j\omega) = A_c \pi [\delta(\omega - \omega_c) e^{j\phi_c} + \delta(\omega + \omega_c) e^{-j\phi_c}] + \frac{A_c}{2} \left[ \underbrace{M_a(j\omega - j\omega_c)}_{\frac{a}{C_M} M(j\omega - j\omega_c)} e^{j\phi_c} + \underbrace{M_a(j\omega + j\omega_c)}_{\frac{a}{C_M} M(j\omega + j\omega_c)} e^{-j\phi_c} \right]$$

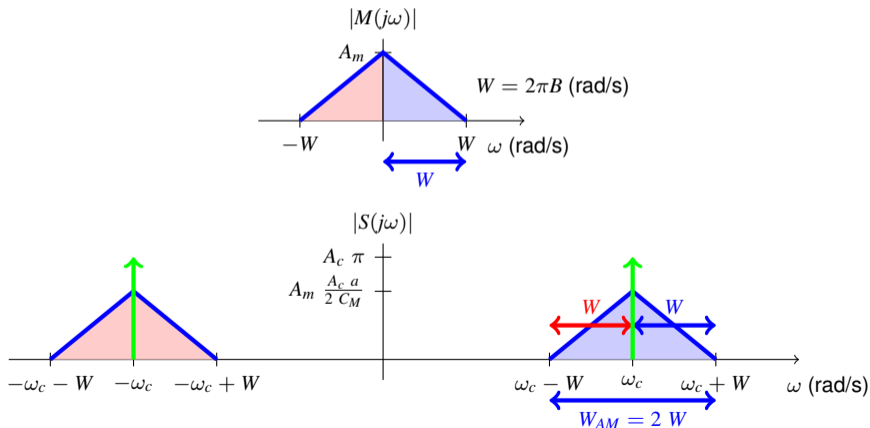
An example:  $m(t)$  with a given shape for  $M(j\omega) = \mathcal{FT}\{m(t)\}$



## Spectrum of the conventional AM signal - Representation (II)

$$S(j\omega) = A_c \pi [\delta(\omega - \omega_c) e^{j\phi_c} + \delta(\omega + \omega_c) e^{-j\phi_c}] + \frac{A_c}{2} \left[ \underbrace{M_a(j\omega - j\omega_c)}_{\frac{a}{C_M} M(j\omega - j\omega_c)} e^{j\phi_c} + \underbrace{M_a(j\omega + j\omega_c)}_{\frac{a}{C_M} M(j\omega + j\omega_c)} e^{-j\phi_c} \right]$$

Another example:  $m(t)$  with another shape for  $M(j\omega) = \mathcal{FT}\{m(t)\}$



## Statistical analysis of conventional AM

- Modulating signal model: random process

$M(t)$  : stationary, with  $m_M = 0$ ,  $R_M(\tau)$ ,  $S_M(j\omega)$ , power  $P_M$  Watts

- ▶ Definition of normalized and with modulation index processes

$$M_n(t) = \frac{1}{C_M} M(t) \quad M_a(t) = a M_n(t) = \frac{a}{C_M} M(t)$$

- Model of the modulated signal: random process

$$S(t) = A_c [1 + M_a(t)] \cos(\omega_c t + \phi_c)$$

- Mean of the conventional AM signal

$$\begin{aligned} m_S(t) = E[S(t)] &= E \left[ \underbrace{A_c [1 + M_a(t)] \cos(\omega_c t + \phi_c)}_{S(t)} \right] \\ &= A_c (1 + E[M_a(t)]) \cos(\omega_c t + \phi_c) \end{aligned}$$

- ▶ If  $M_a(t) = a M_n(t) = \frac{a}{C_M} M(t)$ ,  $E[M_a(t)] = \frac{a}{C_M} E[M(t)] = 0$

$$m_S(t) = A_c \cos(\omega_c t + \phi_c)$$

## Statistical analysis of conventional AM (II)

- Model of the modulated signal: random process

$$S(t) = A_c [1 + M_a(t)] \cos(\omega_c t + \phi_c)$$

- Autocorrelation function of the conventional AM signal

$$\begin{aligned} R_S(t + \tau, t) &= E[S(t + \tau) S(t)] \\ &= E \left[ \left( \underbrace{A_c [1 + M_a(t + \tau)] \cos(\omega_c(t + \tau) + \phi_c)}_{S(t + \tau)} \right) \left( \underbrace{A_c [1 + M_a(t)] \cos(\omega_c t + \phi_c)}_{S(t)} \right) \right] \\ &= A_c^2 E \left[ \underbrace{(1 + M_a(t + \tau))(1 + M_a(t))}_{1 + M_a(t) + M_a(t + \tau) + M_a(t + \tau) M_a(t)} \cos(\omega_c(t + \tau) + \phi_c) \cos(\omega_c t + \phi_c) \right] \end{aligned}$$

$$E[M_a(t)] = E[M_a(t + \tau)] = 0, E[M_a(t + \tau) M_a(t)] = R_{M_a}(\tau), \cos(a) \cos(b) = \frac{1}{2} \cos(a - b) + \frac{1}{2} \cos(a + b)$$

$$R_S(t + \tau, t) = \frac{A_c^2}{2} [1 + R_{M_a}(\tau)] [\cos(\omega_c \tau) + \cos(\omega_c(2t + \tau) + 2\phi_c)]$$



## Statistical analysis of conventional AM (III)

$$m_S(t) = A_c \cos(\omega_c t + \phi_c)$$

$$R_S(t + \tau, t) = \frac{A_c^2}{2} [1 + R_{M_a}(\tau)] [\cos(\omega_c \tau) + \cos(\omega_c(2t + \tau) + 2\phi_c)]$$

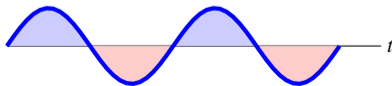
- Process: **Cyclostationary** with period  $T_0 = \frac{2\pi}{\omega_c} = \frac{1}{f_c}$

- ▶ Calculation of the PSD: FT of the time-average of the autocorrelation function

$$S_S(j\omega) = \mathcal{FT}\{\tilde{R}_S(\tau)\}, \text{ with } \tilde{R}_S(\tau) = \frac{1}{T_0} \int_{T_0} R_S(t + \tau, t) dt$$

- ★ Integral of a sinusoid over an integer number of periods

$$\int_{n \times T_0} \cos\left(\frac{2\pi}{T_0} t + \theta\right) dt = \int_{n \times T_0} \sin\left(\frac{2\pi}{T_0} t + \theta\right) dt = 0$$



## Statistical analysis of conventional AM (IV)

- Time average of the autocorrelation function

$$\begin{aligned}\tilde{R}_S(\tau) &= \frac{1}{T_0} \int_{T_0} R_S(t + \tau, t) dt \\ &= \frac{A_c^2}{2} [1 + R_{M_a}(\tau)] \cos(\omega_c \tau) = \frac{A_c^2}{2} \left[ 1 + \frac{a^2}{C_M^2} R_M(\tau) \right] \cos(\omega_c \tau) \\ &= \frac{A_c^2}{2} \cos(\omega_c \tau) + \frac{A_c^2}{2} \frac{a^2}{C_M^2} R_M(\tau) \cos(\omega_c \tau)\end{aligned}$$

$$\text{If } M_a(t) = a M_n(t) = \frac{a}{C_M} M(t), \text{ then } m_{M_a}(t) = a m_{M_n}(t) = \frac{a}{C_M} m_M(t)$$

$$R_{M_a}(\tau) = a^2 R_{M_n}(\tau) = \frac{a^2}{C_M^2} R_M(\tau), \text{ and therefore } S_{M_a}(j\omega) = \frac{a^2}{C_M^2} S_M(j\omega) \text{ and } P_{M_a} = \frac{a^2}{C_M^2} P_M$$

- Power spectral density

$$\begin{aligned}S_S(j\omega) = \mathcal{FT}\{\tilde{R}_S(\tau)\} &= \frac{A_c^2}{2} \pi [\delta(\omega - \omega_c) + \delta(\omega + \omega_c)] \\ &\quad + \frac{A_c^2}{4} [S_{M_a}(j\omega - j\omega_c) + S_{M_a}(j\omega + j\omega_c)]\end{aligned}$$

$$S_S(j\omega) = \frac{A_c^2}{2} \pi [\delta(\omega - \omega_c) + \delta(\omega + \omega_c)] + \frac{A_c^2}{4} \left[ \frac{a^2}{C_M^2} S_M(j\omega - j\omega_c) + \frac{a^2}{C_M^2} S_M(j\omega + j\omega_c) \right]$$

## Statistical analysis of conventional AM (V)

- Power spectral density consists of
  - ▶ Two deltas, in  $-\omega_c$  and in  $+\omega_c$ 
    - ★ Amplitude  $\frac{A_c^2}{2} \pi$
  - ▶ Replicas of  $S_M(j\omega)$  shifted  $-\omega_c$  and  $+\omega_c$ 
    - ★ Scale factor  $\left(\frac{A_c a}{2 C_M}\right)^2$
- Power of the AM modulated signal

$$P_S = \tilde{R}_S(0) = \frac{A_c^2}{2} [1 + R_{M_a}(0)] = \frac{A_c^2}{2} [1 + P_{M_a}] = \frac{A_c^2}{2} \left[ 1 + \frac{a^2}{C_M^2} P_M \right] \text{ Watts}$$

- ▶ Power of the carrier:  $\frac{A_c^2}{2}$  Watts
- ▶ Power of the DSB:  $\left(\frac{A_c^2 a^2}{2 C_M^2}\right) \times P_M$  Watts

NOTE: Power can also be calculated as  $P_S = \frac{1}{2\pi} \int_{-\infty}^{\infty} S_S(j\omega) d\omega$

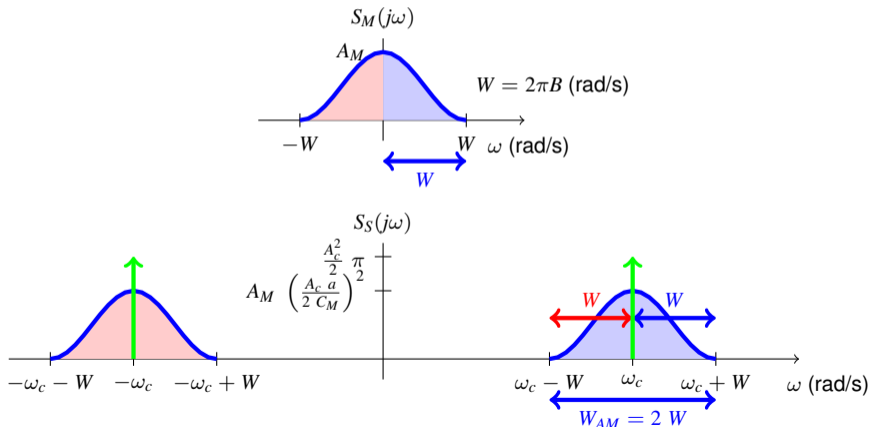
- Bandwidth of the conventional AM signal

$$W_{AM} = 2 W \text{ rad/s}, \quad B_{AM} = 2 B \text{ Hz}$$

## PSD of the conventional AM signal - Representation

$$S_S(j\omega) = \frac{A_c^2}{2} \pi [\delta(\omega - \omega_c) + \delta(\omega + \omega_c)] + \frac{A_c^2}{4} \left[ \frac{a^2}{C_M^2} S_M(j\omega - j\omega_c) + \frac{a^2}{C_M^2} S_M(j\omega + j\omega_c) \right]$$

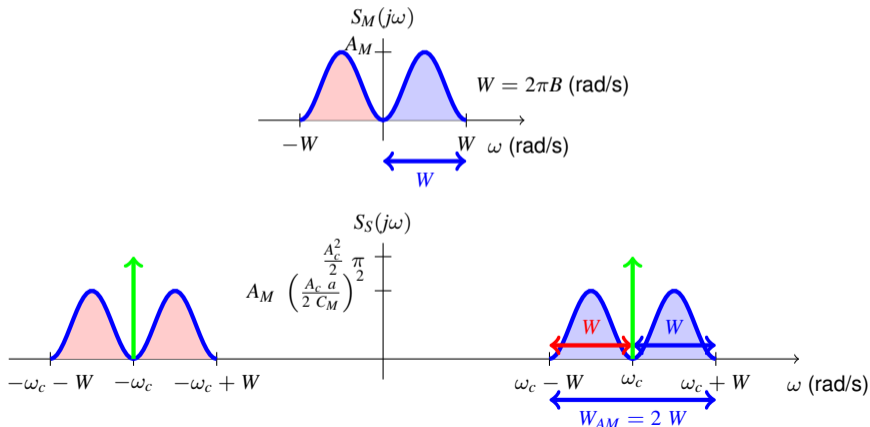
An example: process  $M(t)$  with the following  $S_M(j\omega)$



## PSD of the conventional AM signal - Representation (II)

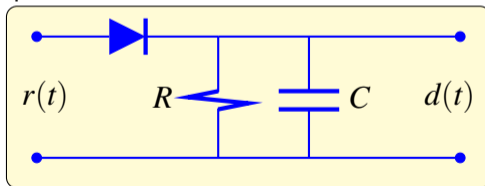
$$S_S(j\omega) = \frac{A_c^2}{2} \pi [\delta(\omega - \omega_c) + \delta(\omega + \omega_c)] + \frac{A_c^2}{4} \left[ \frac{a^2}{C_M^2} S_M(j\omega - j\omega_c) + \frac{a^2}{C_M^2} S_M(j\omega + j\omega_c) \right]$$

Another example: process  $M(t)$  with the following  $S_M(j\omega)$



## Summary of characteristics of conventional AM modulation

- Disadvantages of the conventional AM modulation:
  - ▶ Low power efficiency
    - ★ Power is “wasted” in the transmission of the carrier (which does not contain information)
  - ▶ Low spectral efficiency
    - ★ The bandwidth of the modulated signal is twice that of the modulating signal
- Fundamental advantage of the conventional AM modulation
  - ▶ If  $a \leq 1$ , there is no overmodulation and the signal envelope is proportional to  $1 + m_a(t) \geq 0$ 
    - ★ Recovery of  $m(t)$  : mean subtraction and scaling
  - ▶ Simple receiver: envelope detector



- ★ A synchronous demodulator is not needed (although it can be used as well and is actually the optimal receiver)

## Double Sideband (DSB) modulation (no carrier)

- The carrier of the conventional AM modulation is suppressed
  - ▶ Eliminates the power efficiency drawback of conventional AM

$$s(t) = m(t) \times c(t) = m(t) \times A_c \cos(\omega_c t + \phi_c)$$

$$\text{Conv-AM: } s(t) = c(t) + \frac{a}{C_M} \times m(t) \times c(t)$$

- Frequency response (deterministic signal  $m(t)$  with  $M(j\omega) = \mathcal{FT}\{m(t)\}$ )

$$\begin{aligned} S(j\omega) &= \frac{1}{2\pi} \mathcal{FT}\{m(t)\} * \mathcal{FT}\{A_c \cos(\omega_c t + \phi_c)\} \\ &= \frac{A_c}{2} [M(j\omega - j\omega_c) e^{j\phi_c} + M(j\omega + j\omega_c) e^{-j\phi_c}] \end{aligned}$$

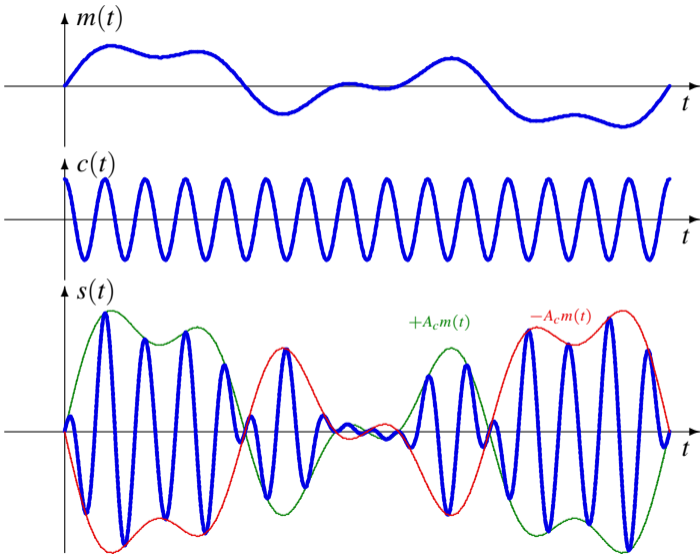
- ▶ The deltas of conventional AM modulation disappear
- ▶ Replicas of  $M(j\omega)$  shifted  $\pm\omega_c$  rad/s
  - ★ The scaling of the replicas is simpler (since there is no normalization)
  - ★ Name: two sidebands, lower ( $|\omega| < \omega_c$ ) and upper ( $|\omega| > \omega_c$ )

- Bandwidth

$$W_{DSB} = 2 W \text{ rad/s, } B_{DSB} = 2 B \text{ Hz}$$

It is still twice that of the modulating signal being transmitted

# Waveform of a DSB modulation

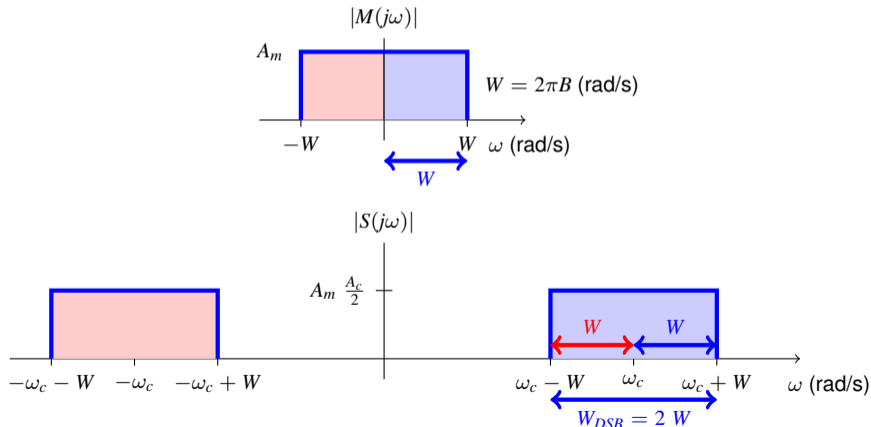




## Spectrum of the DSB signal - Representation

$$S(j\omega) = \frac{A_c}{2} [M(j\omega - j\omega_c) e^{j\phi_c} + M(j\omega + j\omega_c) e^{-j\phi_c}]$$

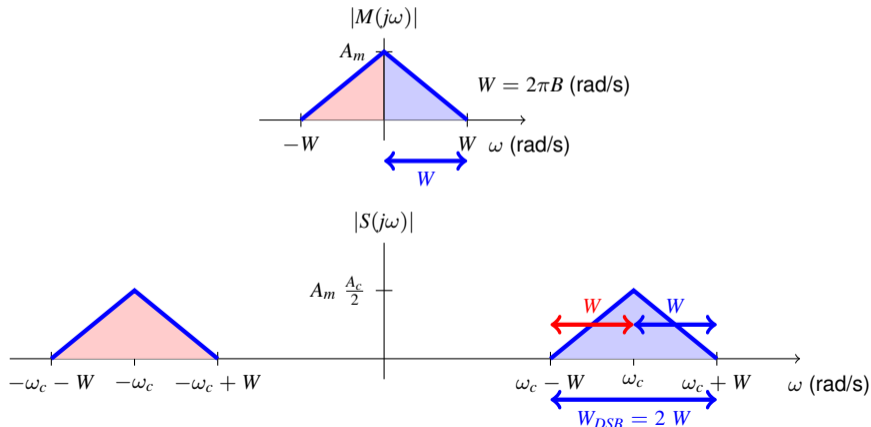
An example:  $m(t)$  with a given shape for  $M(j\omega) = \mathcal{FT}\{m(t)\}$



## Spectrum of the DSB signal - Representation (II)

$$S(j\omega) = \frac{A_c}{2} [M(j\omega - j\omega_c) e^{j\phi_c} + M(j\omega + j\omega_c) e^{-j\phi_c}]$$

Another example:  $m(t)$  with another shape for  $M(j\omega) = \mathcal{FT}\{m(t)\}$



## Statistical analysis of the DSB modulation

- Modulating signal model: random process

$M(t)$  : stationary, with  $m_M = 0$ ,  $R_M(\tau)$ ,  $S_M(j\omega)$  and power  $P_M$  Watts

- Model of the modulated signal: random process

$$S(t) = M(t) \times c(t) = A_c M(t) \cos(\omega_c t + \phi_c)$$

- Mean of the DSB modulated signal

$$m_S(t) = E[S(t)] = A_c E[M(t)] \cos(\omega_c t + \phi_c) = 0$$

- Autocorrelation function of the DSB modulated signal

$$\begin{aligned} R_S(t + \tau, t) &= E[S(t + \tau)S(t)] \\ &= A_c^2 E[M(t + \tau) M(t)] \cos(\omega_c(t + \tau) + \phi_c) \cos(\omega_c t + \phi_c) \\ &= \frac{A_c^2}{2} R_M(\tau) [\cos(\omega_c \tau) + \cos(\omega_c(2t + \tau) + 2\phi_c)] \end{aligned}$$

$$\cos(a) \cos(b) = \frac{1}{2} \cos(a - b) + \frac{1}{2} \cos(a + b)$$

- Process: **Cyclostationary** with period  $T_0 = \frac{2\pi}{2\omega_c} = \frac{1}{2f_c}$

## Statistical analysis of the DSB modulation (II)

- Time average of the autocorrelation function

$$\tilde{R}_S(\tau) = \frac{1}{T} \int_{T_0} R_S(t + \tau, t) dt = \frac{A_c^2}{2} R_M(\tau) \cos(\omega_c \tau)$$

- Power spectral density

$$S_S(j\omega) = \mathcal{FT}\{\tilde{R}_S(\tau)\} = \frac{A_c^2}{4} [S_M(j\omega - j\omega_c) + S_M(j\omega + j\omega_c)]$$

- DSB signal power

$$P_S = \tilde{R}_S(0) = \frac{A_c^2}{2} R_M(0) = \frac{A_c^2}{2} P_M$$

- ▶ Power efficient

★ Carrierless : no power is "wasted" on terms that contain no information

- DSB modulation bandwidth

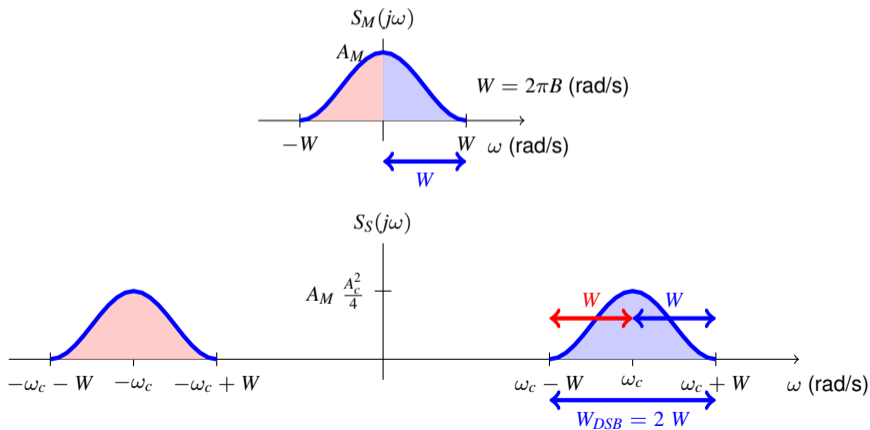
$$W_{DSB} = 2 W \text{ rad/s}, B_{DSB} = 2 B \text{ Hz}$$

- ▶ It is still twice that of the modulating signal

## PSD of the DSB signal - Representation

$$S_S(j\omega) = \frac{A_c^2}{4} [S_M(j\omega - j\omega_c) + S_M(j\omega + j\omega_c)]$$

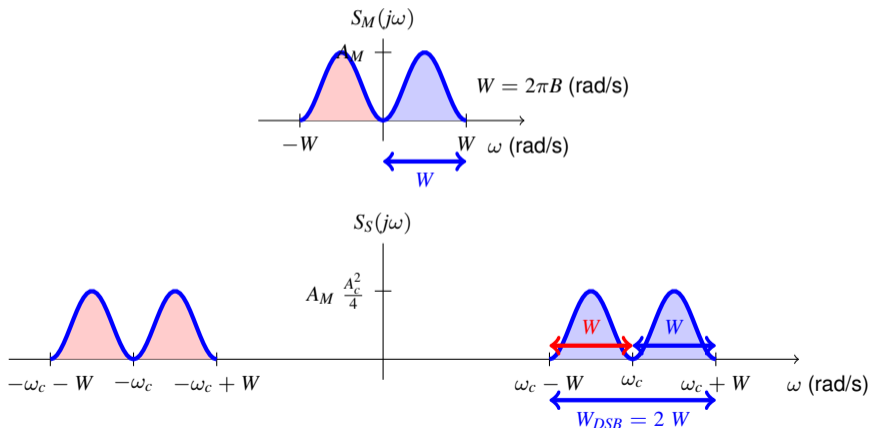
An example: process  $M(t)$  with the following  $S_M(j\omega)$



## PSD of the DSB signal - Representation (II)

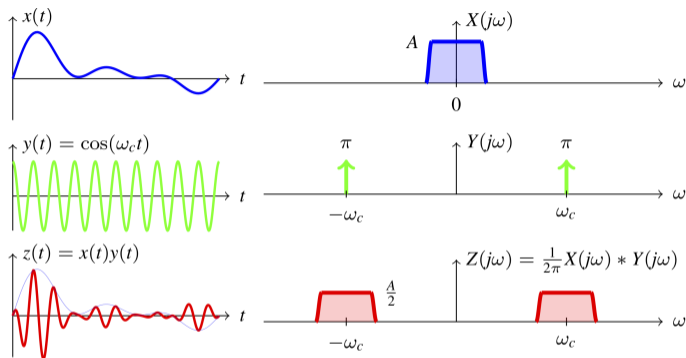
$$S_S(j\omega) = \frac{A_c^2}{4} [S_M(j\omega - j\omega_c) + S_M(j\omega + j\omega_c)]$$

Another example: process  $M(t)$  with the following  $S_M(j\omega)$



## Revision : Product by a sinusoid - Effect in the frequency domain

- Sinusoid of frequency  $\omega_c$ : two replicas of the spectrum of the modulated signal, shifted  $\pm\omega_c$  rad/s



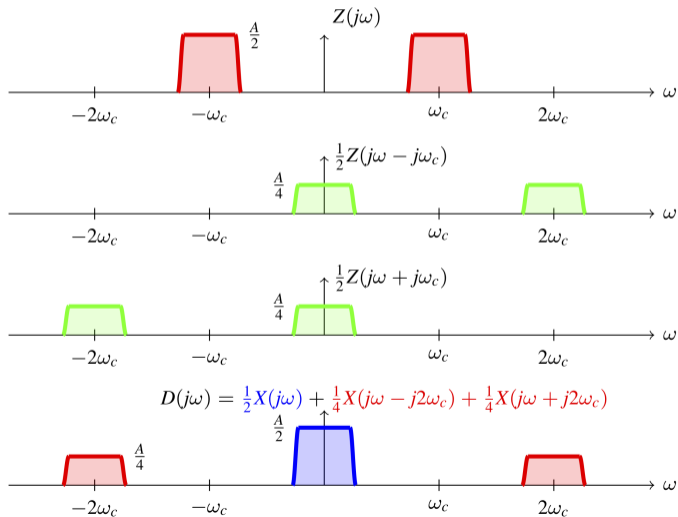
$$z(t) = x(t) \cos(\omega_c t) \stackrel{\mathcal{FT}}{\leftrightarrow} Z(j\omega) = \frac{1}{2} X(j\omega - j\omega_c) + \frac{1}{2} X(j\omega + j\omega_c)$$

- Power spectral density (for random signals)

$$S_Z(j\omega) = \frac{1}{4} S_X(j\omega - j\omega_c) + \frac{1}{4} S_X(j\omega + j\omega_c)$$

## Product by a sinusoid - Effect on frequency (II)

$$d(t) = z(t) \cos(\omega_c t) \xleftrightarrow{\mathcal{F}\mathcal{T}} D(j\omega) = \frac{1}{2}Z(j\omega - j\omega_c) + \frac{1}{2}Z(j\omega + j\omega_c)$$

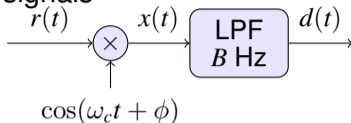


$$D(j\omega) = \frac{1}{2}Z(j\omega) + \frac{1}{4}Z(j\omega - j2\omega_c) + \frac{1}{4}Z(j\omega + j2\omega_c)$$



## DSB demodulation: Synchronous or coherent receiver

- Receiver for DSB modulated signals



LPF: low pass filter (with bandwidth  $B$  Hz)

- ▶ Demodulator (product with the carrier  $\cos(\omega_c t + \phi)$ )
- ▶ Low pass filter (bandwidth given by the signal,  $B$  Hz)
- Optimum performance with a synchronous or coherent receiver
  - ▶ Same phase in the receiver carrier as in the transmitter carrier

$$\phi = \phi_c$$

- Effect of a non-synchronous receiver ( $\phi \neq \phi_c$ )
  - ▶ Attenuation of the term related to the signal  $m(t)$
  - ▶ Loss of signal to noise ratio (performance)
    - ★ The value of the phase  $\phi$  does not vary the power due to the noise term

## Effect of a non-coherent receiver

- Analysis of the signal term  $r(t) = s(t) = A_c m(t) \cos(\omega_c t + \phi_c)$
- Unfiltered demodulated signal

$$\begin{aligned}x(t) &= r(t) \times \cos(\omega_c t + \phi) \\ &= A_c m(t) \cos(\omega_c t + \phi_c) \cos(\omega_c t + \phi) \\ &= \frac{A_c}{2} m(t) [\cos(\phi - \phi_c) + \cos(2\omega_c t + \phi_c + \phi)]\end{aligned}$$

- Filtered demodulated signal
  - ▶ Terms with spectrum in  $\pm 2\omega_c$  are eliminated

$$d(t) = \frac{A_c}{2} m(t) \cos(\phi - \phi_c)$$

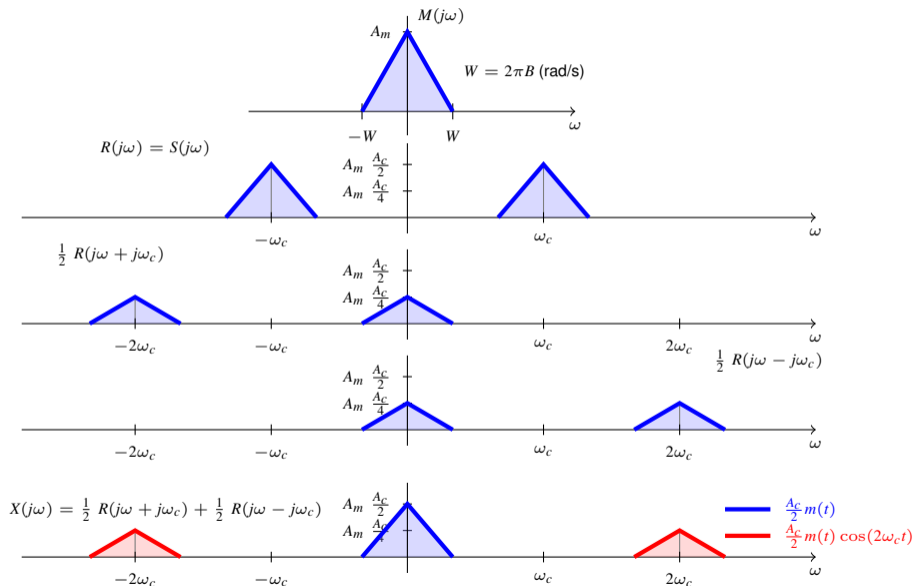
- ▶ Ideal value, with coherent receiver ( $\phi = \phi_c$ )

$$d(t) = \frac{A_c}{2} m(t)$$

- ▶ Effect of using a non-coherent receiver ( $\phi \neq \phi_c$ )
  - ★ Attenuation Term

$$\cos(\phi - \phi_c)$$

# Synchronous Demodulation of DSB - Frequency Interpretation

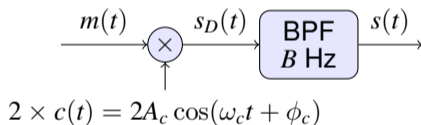


## Coherent Receiver - Possible Options

- The receiver must identify the phase of the carrier with which the signal was modulated  $\phi = \phi_c$
- Frequent options
  - ▶ Transmission of a pilot (carrier of reduced amplitude)
    - ★ Power inefficiency
  - ▶ Use of a phase-locked loop (PLL: *Phase Locked Loop*)
    - ★ Increases the cost of the receiver

# Single SideBand (SSB) modulation

- Spectral efficiency:  $B_{SSB} = B$  Hz
  - ▶ One sideband is removed
- Signal generation by direct filtering
  - ▶ A double sideband signal is generated (with double amplitude)
  - ▶ One of the two sidebands is removed by filtering
    - ★ SSB Upper SideBand (USB): frequencies  $|\omega| < \omega_c$  are removed
    - ★ SSB Lower SideBand (LSB): frequencies  $|\omega| > \omega_c$  are removed



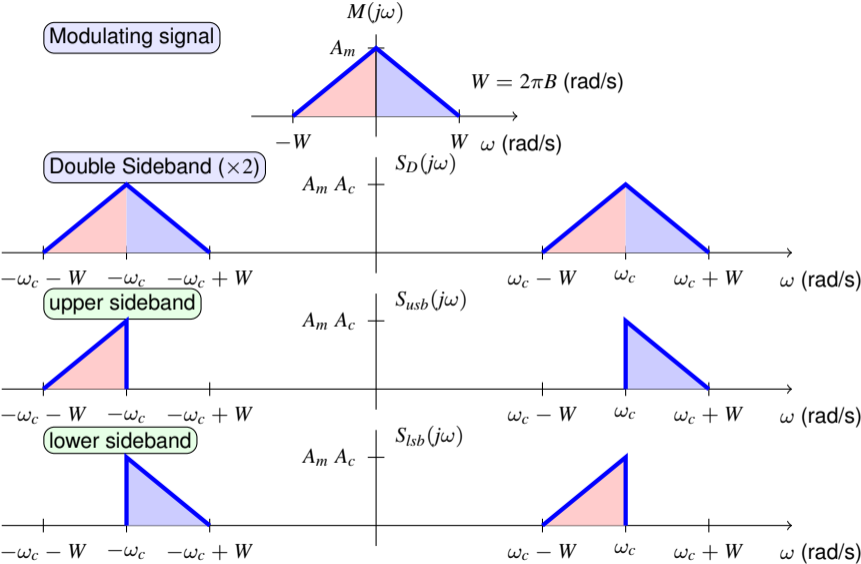
- Analytical expression of the resulting SSB signal

$$s(t) = A_c m(t) \cos(\omega_c t + \phi_c) \mp A_c \hat{m}(t) \sin(\omega_c t + \phi_c)$$

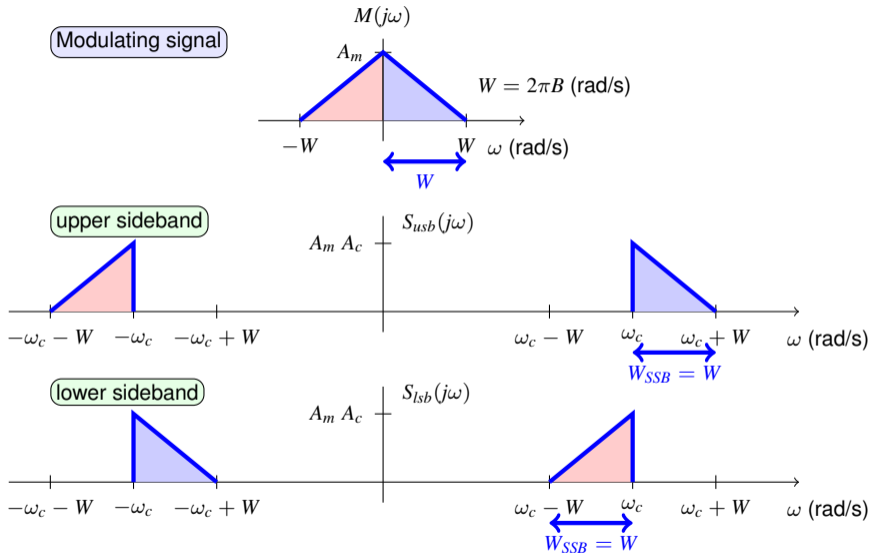
$\hat{m}(t)$ : Hilbert transform of the modulating signal  $m(t)$

- ▶ Upper sideband (USB):  $-$  sign
- ▶ Lower sideband (LSB):  $+$  sign

# Spectrum of the SSB signal



# SSB AM signal spectrum - Bandwidth



## Hilbert transform

- Signal generated by filtering with a Hilbert transformer

$$\hat{m}(t) = m(t) * h_{Hilbert}(t)$$

- Hilbert transformer:

- ▶ Impulse response

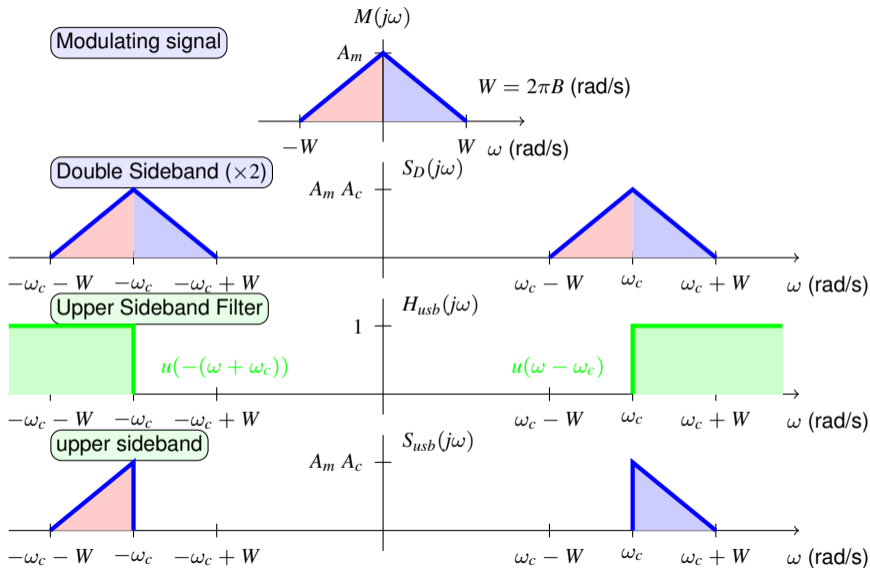
$$h_{Hilbert}(t) = \frac{1}{\pi t}$$

- ▶ Frequency response

$$H_{Hilbert}(j\omega) = \begin{cases} -j, & \omega > 0 \\ +j, & \omega < 0 \\ 0, & \omega = 0 \end{cases}$$



# Upper Sideband SSB Generation



## Analytic expression of $s(t)$ - Upper sideband

- Frequency response of the upper sideband filter:

$$H_{usb}(j\omega) = u(\omega - \omega_c) + u(-(\omega + \omega_c)) \text{ with } u(x) \text{ step function}$$

- Spectrum of the DBL signal with double amplitude,  $s_D(t)$

$$S_D(j\omega) = A_c [M(j\omega - j\omega_c) + M(j\omega + j\omega_c)]$$

- SSB signal spectrum with upper sideband

$$\begin{aligned} S(j\omega) &= S_D(j\omega) H_{usb}(j\omega) \\ &= A_c M(j\omega) u(\omega) |_{\omega=\omega-\omega_c} + A_c M(j\omega) u(-\omega) |_{\omega=\omega+\omega_c} \end{aligned}$$

The following properties of the Fourier transform and Euler's formulas for sinusoids are used

$$\mathcal{FT} \left\{ \frac{1}{2} \delta(t) + \frac{j}{2\pi t} \right\} = u(\omega), \quad \mathcal{FT} \left\{ \frac{1}{2} \delta(t) - \frac{j}{2\pi t} \right\} = u(-\omega), \quad \mathcal{FT} \{x(t) e^{j\omega_c t}\} = X(j\omega - j\omega_c)$$

$$\cos(\omega_c t) = \frac{e^{+j\omega_c t} + e^{-j\omega_c t}}{2}, \quad \sin(\omega_c t) = \frac{e^{+j\omega_c t} - e^{-j\omega_c t}}{2j} = j \frac{e^{-j\omega_c t} - e^{+j\omega_c t}}{2}$$

- Upper sideband SSB signal

$$\begin{aligned} s_{usb}(t) &= A_c m(t) * \left[ \frac{1}{2} \delta(t) + \frac{j}{2\pi t} \right] e^{j\omega_c t} + A_c m(t) * \left[ \frac{1}{2} \delta(t) - \frac{j}{2\pi t} \right] e^{-j\omega_c t} \\ &= \frac{A_c}{2} [m(t) + j\hat{m}(t)] e^{j\omega_c t} + \frac{A_c}{2} [m(t) - j\hat{m}(t)] e^{-j\omega_c t} \\ &= A_c m(t) \cos(\omega_c t) - A_c \hat{m}(t) \sin(\omega_c t) \end{aligned}$$

## Analytic expression of $s(t)$ - Lower sideband

- Upper sideband SSB signal

$$s_{usb}(t) = A_c m(t) \cos(\omega_c t) - A_c \hat{m}(t) \sin(\omega_c t)$$

- Relations of the two SSB signals and signal  $s_D(t)$

$$s_D(t) = 2 A_c m(t) \cos(\omega_c t) = s_{usb}(t) + s_{lsb}(t)$$

- Lower sideband SSB signal

$$s_{lsb}(t) = s_D(t) - s_{usb}(t)$$

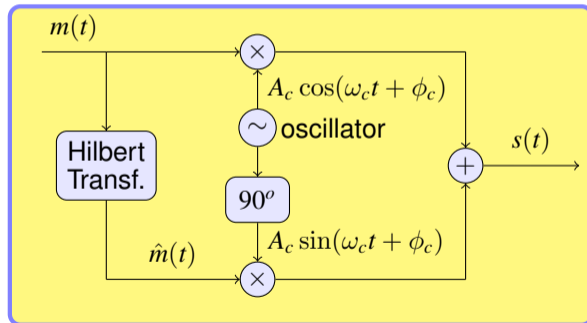
$$s_{lsb}(t) = A_c m(t) \cos(\omega_c t) + A_c \hat{m}(t) \sin(\omega_c t)$$

## Alternative SSB Generation - Hartley Modulator

- Implementation based on the analytic expression from the Hilbert transform

$$s(t) = A_c m(t) \cos(\omega_c t + \phi_c) \mp A_c \hat{m}(t) \sin(\omega_c t + \phi_c)$$

REMARK: for ease of notation, we had previously considered  $\phi_c = 0$ , now an arbitrary value is used



## Bandwidth and power of the SSB signal

- Power spectral density
  - ▶ upper sideband

$$S_{S_{usb}}(j\omega) = \begin{cases} A_c^2 [S_M(j\omega - j\omega_c) + S_M(j\omega + j\omega_c)], & |\omega| > \omega_c \\ 0, & |\omega| < \omega_c \end{cases}$$

- ▶ lower sideband

$$S_{S_{lsb}}(j\omega) = \begin{cases} 0, & |\omega| > \omega_c \\ A_c^2 [S_M(j\omega - j\omega_c) + S_M(j\omega + j\omega_c)], & |\omega| < \omega_c \end{cases}$$

- Signal power

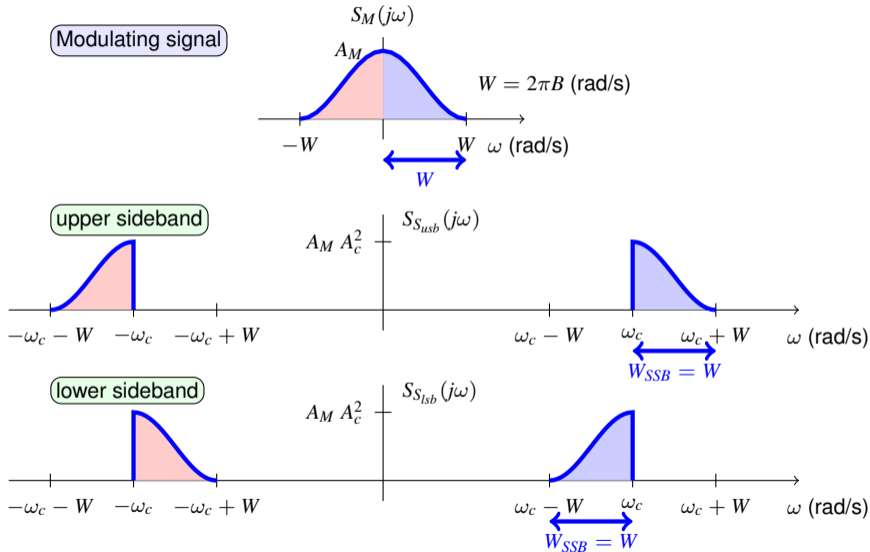
$$P_S = \frac{1}{2\pi} \int_{-\infty}^{\infty} S_S(j\omega) d\omega = A_c^2 P_M \text{ Watts}$$

- Bandwidth

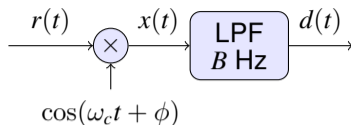
$$W_{SSB} = W \text{ rad/s}, B_{SSB} = B \text{ Hz}$$

Same bandwidth as the transmitted modulating signal

# PSD of the SSB signal - Representation



## Coherent (synchronous) demodulation of SSB signals



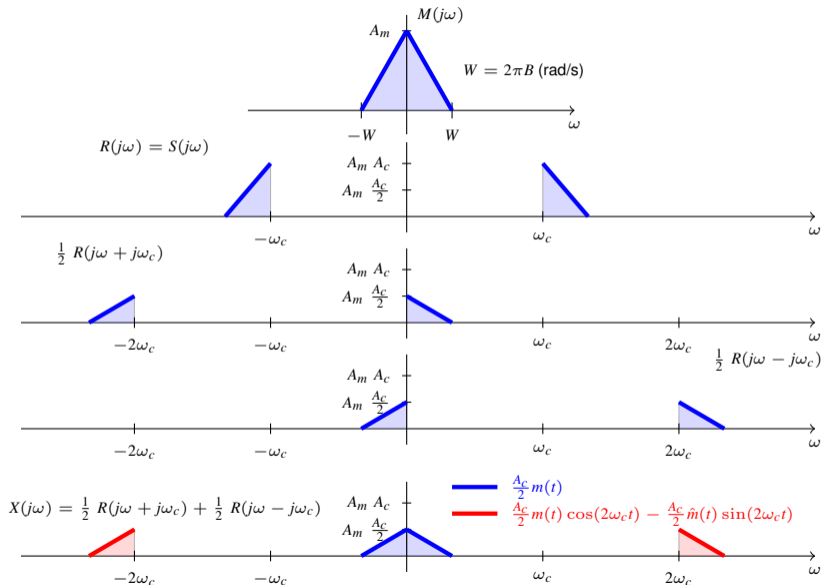
- Received signal  $r(t) = s(t) = A_c m(t) \cos(\omega_c t + \phi_c) \mp A_c \hat{m}(t) \sin(\omega_c t + \phi_c)$
- Unfiltered demodulated signal  $x(t)$

$$\begin{aligned}x(t) &= r(t) \times \cos(\omega_c t + \phi) \\&= [A_c m(t) \cos(\omega_c t + \phi_c) \mp A_c \hat{m}(t) \sin(\omega_c t + \phi_c)] \times \cos(\omega_c t + \phi) \\&= \frac{1}{2} A_c m(t) \cos(\phi - \phi_c) \pm \frac{1}{2} A_c \hat{m}(t) \sin(\phi - \phi_c) \\&\quad + \frac{1}{2} A_c m(t) \cos(2\omega_c t + \phi + \phi_c) \mp \frac{1}{2} A_c \hat{m}(t) \sin(2\omega_c t + \phi + \phi_c)\end{aligned}$$

- Filtered demodulated signal

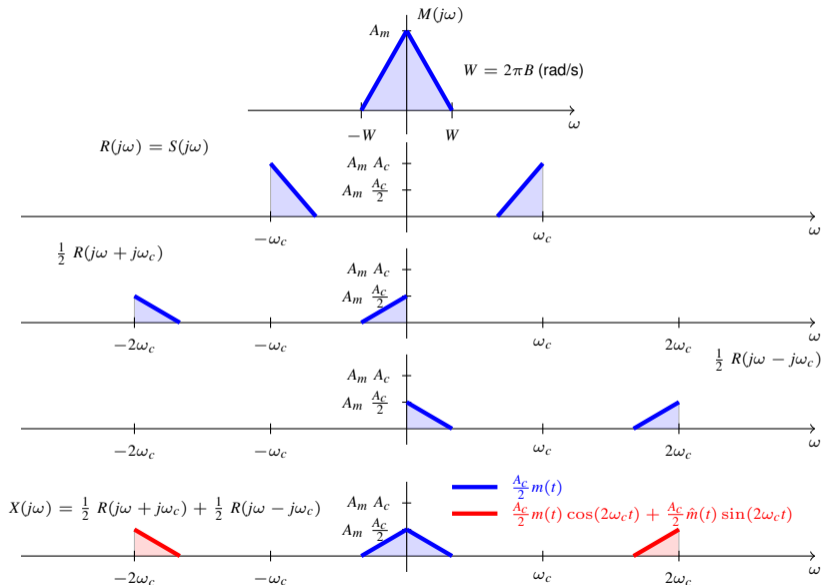
$$d(t) = \frac{1}{2} A_c m(t) \cos(\phi - \phi_c) \pm \frac{1}{2} A_c \hat{m}(t) \sin(\phi - \phi_c)$$

# Synchronous Demodulation of SSB (USB) - Frequency Interpretation





# Synchronous Demodulation of SSB (LSB) - Frequency Interpretation



## Demodulation of SSB signals (II)

- Filtered demodulated signal (with arbitrary non-zero  $\phi_c$  phase)

$$d(t) = \frac{1}{2} A_c m(t) \cos(\phi - \phi_c) \pm \frac{1}{2} A_c \hat{m}(t) \sin(\phi - \phi_c)$$

- Negative effects present with non-coherent demodulators

- ▶ Attenuation of the received signal term due to  $m(t)$

$$\frac{A_c}{2} m(t) \cos(\phi - \phi_c), \text{ attenuation term } \cos(\phi - \phi_c)$$

- ★ Same as for double sideband modulation

- ▶ Additional distortion term

$$\pm \frac{A_c}{2} \hat{m}(t) \sin(\phi - \phi_c), \text{ gain term } \sin(\phi - \phi_c)$$

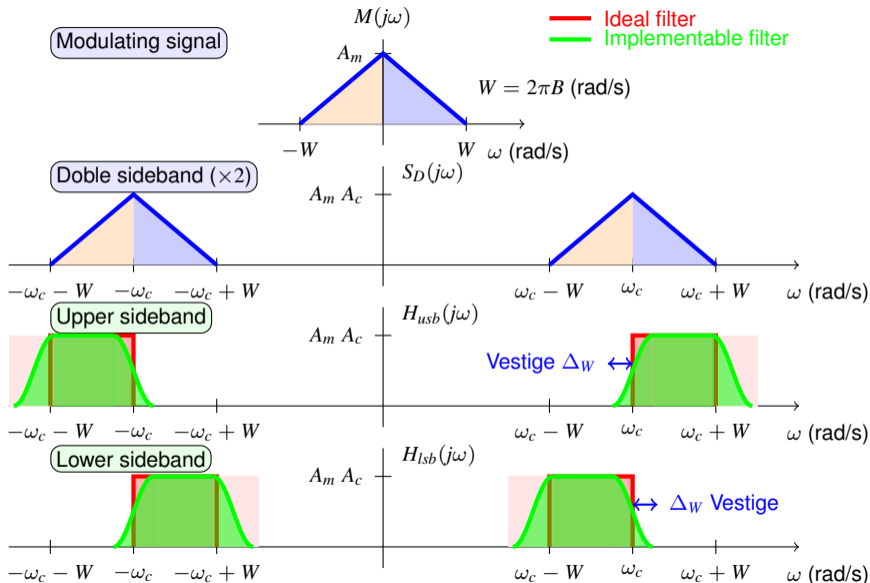
- ★ Situation worse than for double sideband modulation

- Need for a synchronous or coherent demodulator

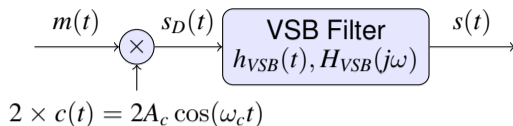
## Characteristics of single sideband modulation

- SSB modulation overcomes the two main drawbacks of conventional AM modulation
  - ▶ Spectral efficiency
    - ★ Same bandwidth as the information signal that is transmitted
  - ▶ Power efficiency
    - ★ All the power of the signal is associated with the component that contains the information (no energy is used to transmit a carrier)
- Disadvantage of SSB modulation
  - ▶ Implementation using direct filtering: requires ideal filters to eliminate one of the sidebands
    - ★ The implementation with real filters can generate distortion in the transmitted signal
  - ▶ Implementation with Hartley modulator: requires a Hilbert transformer
    - ★ Ideal response of the transformer (not achievable without error)

# SSB to VSB : Form ideal filters to implmentable filters



## Vestigial SideBand (VSB) modulation



- Same modulation scheme as SSB
  - ▶ The VSB “*ideal*” filter is replaced by an implementable filter
    - ★ Vestigial sideband filter (which must meet certain conditions)
- VSB modulated signal
  - ▶ A double amplitude double sideband signal  $s_D(t)$  is filtered with a VSB filter

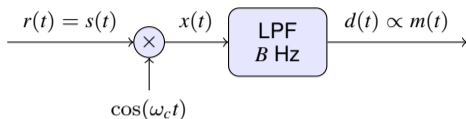
$$s(t) = \left[ \underbrace{m(t) \times 2A_c \cos(\omega_c t)}_{s_D(t)} \right] * h_{VSB}(t)$$

- VSB signal in the frequency domain

$$S(j\omega) = A_c [M(j\omega - j\omega_c) + M(j\omega + \omega_c)] H_{VSB}(j\omega)$$

## Characteristics of the VSB filter

- The received signal and its dependence on the filter will be analyzed.
  - ▶ Obtaining the conditions that must be met



- Received signal (same as transmitted) in the frequency domain

$$R(j\omega) = S(j\omega) = A_c [M(j\omega - j\omega_c) + M(j\omega + \omega_c)] H_{VSB}(j\omega)$$

- Demodulated (unfiltered) signal in the frequency domain

$$x(t) = r(t) \cos(\omega_c t) \rightarrow X(j\omega) = \frac{1}{2} [R(j\omega - j\omega_c) + R(j\omega + j\omega_c)]$$

$$X(j\omega) = \frac{A_c}{2} [M(j\omega - j2\omega_c) + M(j\omega)] H_{VSB}(j\omega - j\omega_c) + \frac{A_c}{2} [M(j\omega) + M(j\omega + j2\omega_c)] H_{VSB}(j\omega + j\omega_c)$$

- Filtered demodulated signal in the frequency domain

$$D(j\omega) = \frac{A_c}{2} M(j\omega) [H_{VSB}(j\omega - j\omega_c) + H_{VSB}(j\omega + j\omega_c)]$$

## Characteristics of the VSB filter (II)

- Filtered demodulated signal in the frequency domain

$$D(j\omega) = \frac{A_c}{2} M(j\omega) [H_{VSB}(j\omega - j\omega_c) + H_{VSB}(j\omega + j\omega_c)]$$

- ▶ Interpretation: the modulating signal is filtered with the equivalent filter

$$H_{EQ}(j\omega) = H_{VSB}(j\omega - j\omega_c) + H_{VSB}(j\omega + j\omega_c)$$

- The filter must have a response that in  $|\omega| \leq 2\pi B$  rad/s meets

- ▶ Constant module
- ▶ Linear phase

- Therefore, the conditions that the VSB filter must meet are

$$|H_{VSB}(j\omega - j\omega_c) + H_{VSB}(j\omega + j\omega_c)| = C, \text{ in } |\omega| \leq 2\pi B \text{ rad/s}$$

$$\text{Odd symmetry around } \omega_c \text{ in } \omega_c - \Delta_W < \omega < \omega_c + \Delta_W \text{ rad/s}$$

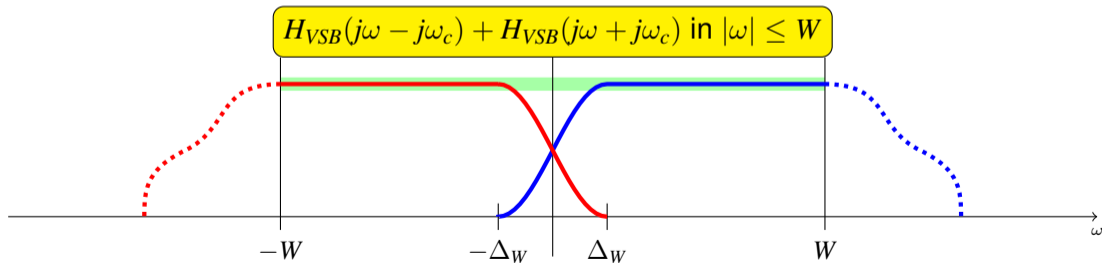
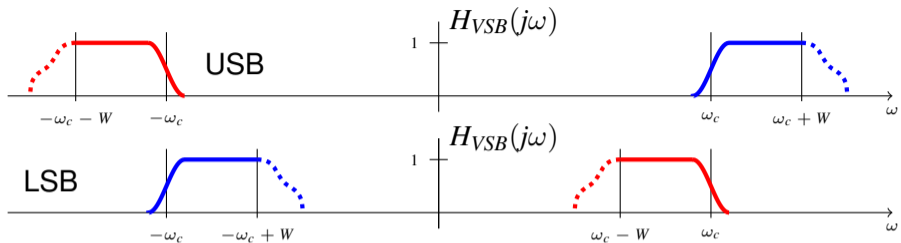
$\Delta_W$ : bandwidth excess (vestige) in rad/s

- Bandwidth of the modulated signals

$$W_{VSB} = W + \Delta_W \text{ rad/s, or } B_{VSB} = B + \Delta_B \text{ Hz, with } \Delta_B = \frac{\Delta_W}{2\pi}$$

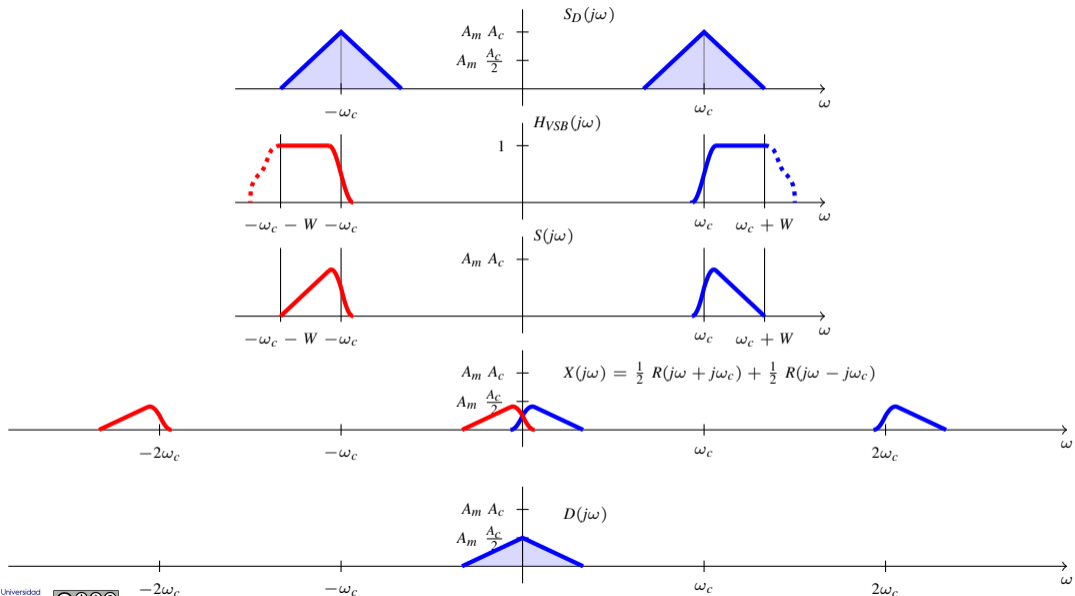
$\Delta_B$ : bandwidth excess of VSB in Hz (typically  $\Delta_B \ll B$ )

# VSB Filter

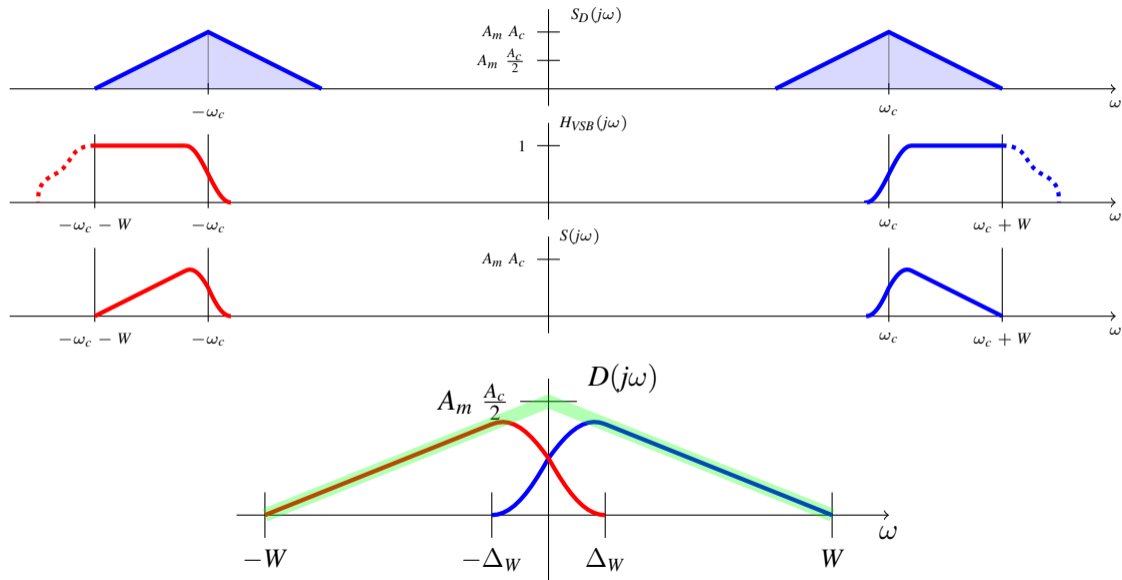




# Synchronous demodulation of the VSB (USB) - Frequency Interpretation



# Synchronous demodulation of the VSB (USB) - Frequency Interpretation



# Amplitude modulations - Summary

Modulation	$BW$ (Hz)	$P_S$	$P_S(m(t))$	$d(t)$	$P_d(m(t))$
Conventional AM	$2B$	$\frac{A_c^2}{2} [1 + P_{M_a}]$	$\frac{A_c^2}{2} P_{M_a}$	$\frac{A_c}{2} [1 + m_a(t)]$	$\frac{A_c^2}{4} P_{M_a}$
DSB	$2B$	$\frac{A_c^2}{2} P_M$	$\frac{A_c^2}{2} P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$
SSB	$B$	$A_c^2 P_M$	$A_c^2 P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$
VSB	$B + \Delta_B$	$A_c^2 P_M$	$A_c^2 P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$

$BW$  (Hz) : bandwidth of the modulated signal in Hz

$P_S$  : power of the modulated signal

$P_S(m(t))$  : power of the modulated signal relative to  $m(t)$

$d(t)$  : signal recovered with a synchronous or coherent receiver

$P_d(m(t))$  : power of the demodulated signal relative to  $m(t)$

- Power efficiency

- ▶ All signal power is related to  $m(t)$

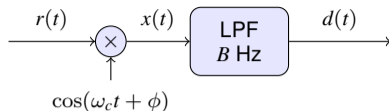
- ★ DSB, SSB y VSB

- Spectral efficiency

- ▶ Minimum transmission bandwidth (same bandwidth as the modulating signal,  $B$  Hz)

- ★ SSB and VSB (in this case with a vestigial increment  $\Delta_B$ )

# Synchronous demodulation of conventional AM modulation



- Received signal: Conv. AM  $r(t) = s(t) = A_c [1 + m_a(t)] \cos(\omega_c t + \phi_c)$
- Unfiltered demodulated signal  $x(t)$

$$\begin{aligned}x(t) &= r(t) \times \cos(\omega_c t + \phi) \\ &= A_c [1 + m_a(t)] \cos(\omega_c t + \phi_c) \times \cos(\omega_c t + \phi) \\ &= \frac{A_c}{2} [1 + m_a(t)] \cos(\phi_c - \phi) + \frac{A_c}{2} [1 + m_a(t)] \cos(2\omega_c t + \phi_c + \phi)\end{aligned}$$

- Filtered demodulated signal

$$d(t) = \frac{A_c}{2} [1 + m_a(t)] \cos(\phi_c - \phi)$$

- Need for a synchronous or coherent demodulator:  $\phi = \phi_c$

$$d(t) = \frac{A_c}{2} [1 + m_a(t)]$$

# ANGLE MODULATIONS

## PM AND FM

## Angle modulations

- The information is in the argument of a sinusoidal carrier

$$c(t) = A_c \cos(2\pi f_c t + \phi_c)$$

It is printed in the phase or frequency of the carrier

- ▶ Phase Modulation (PM)

$$\phi_c \rightarrow \phi_c(t) = f(m(t))$$

- ▶ Frequency modulation (FM)

$$f_i(t) = f_c \rightarrow f_i(t) = f(m(t))$$

$f_i(t)$ : instantaneous frequency of the carrier signal

- Common representation of FM and PM signals

$$s(t) = A_c \cos(\theta(t))$$

The information is in

$$\theta(t) = f(m(t))$$

## Instantaneous frequency

- Common representation of PM and FM signals

$$s(t) = A_c \cos(\theta(t))$$

- ▶ The information is in the phase term

$$\theta(t) = \omega_c t + \phi(t) \text{ rad/s}$$

- Definition of instantaneous frequency of a sinusoid

$$\omega_i(t) = \frac{d}{dt}\theta(t) \text{ rad/s}, \quad f_i(t) = \frac{1}{2\pi} \frac{d}{dt}\theta(t) \text{ Hz}$$

- Modulated signal  $s(t)$  and instantaneous frequency

$$s(t) = A_c \cos(\underbrace{\omega_c t}_{2\pi f_c t} + \phi(t)), \quad f_i(t) = f_c + \frac{1}{2\pi} \frac{d}{dt}\phi(t) \text{ Hz}$$

## Phase (PM) and Frequency (FM) modulations

- Modulated signal  $s(t)$  and instantaneous frequency

$$s(t) = A_c \cos(\underbrace{\omega_c t}_{2\pi f_c t} + \phi(t)), \quad f_i(t) = f_c + \frac{1}{2\pi} \frac{d}{dt} \phi(t) \text{ Hz}$$

- If  $m(t)$  is the modulating signal (message)

- ▶ Phase Modulation (PM)

$$\phi(t) = k_p m(t) \text{ rad/s}$$

★  $k_p$ : phase deviation constant

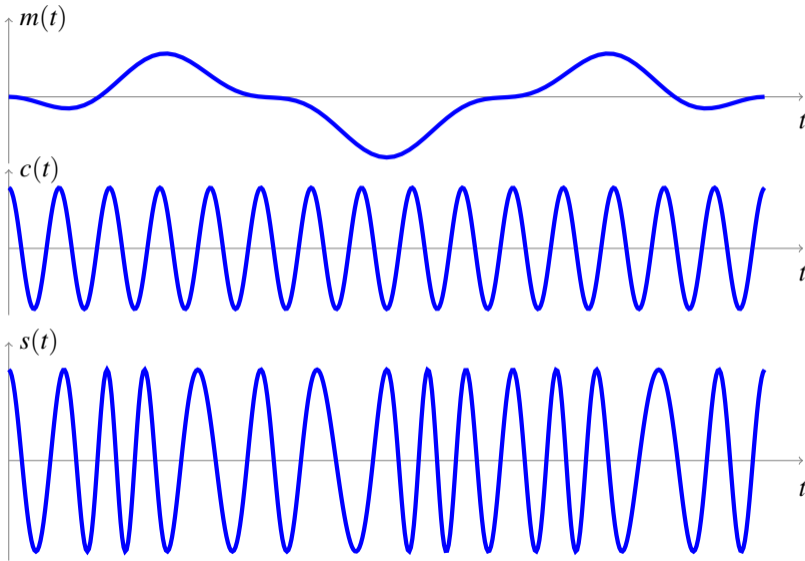
- ▶ Frequency Modulation (FM)

$$\Delta f_i(t) = f_i(t) - f_c = \frac{1}{2\pi} \frac{d}{dt} \phi(t) = k_f m(t) \text{ Hz}$$

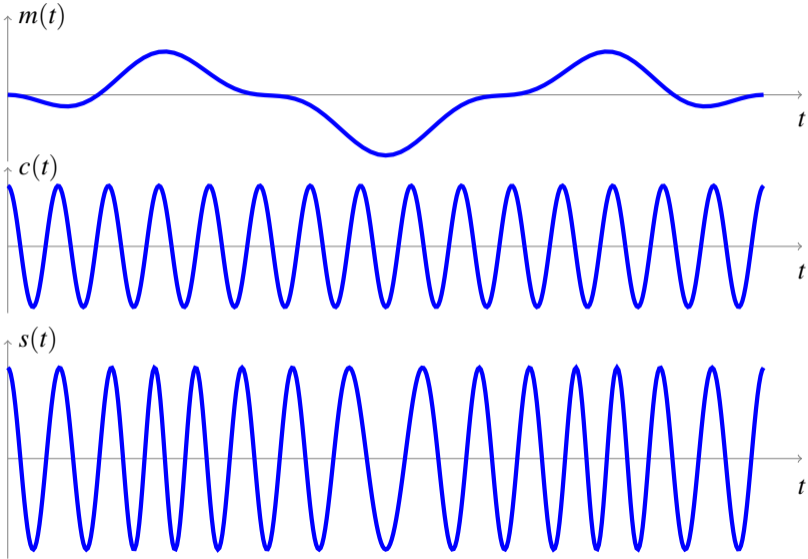
★  $k_f$ : frequency deviation constant



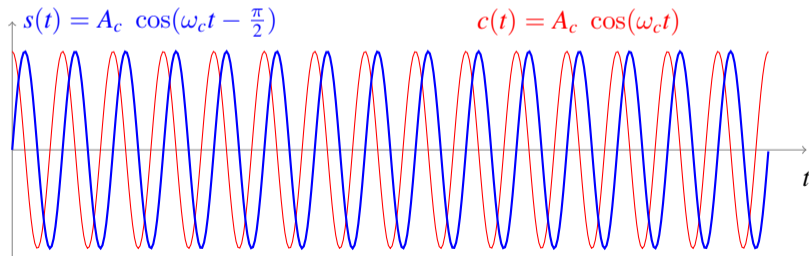
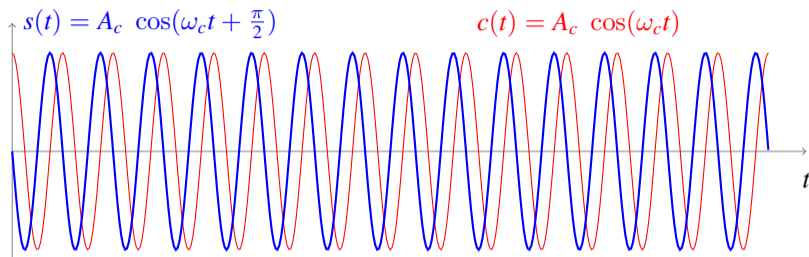
# Waveform of a PM Modulation



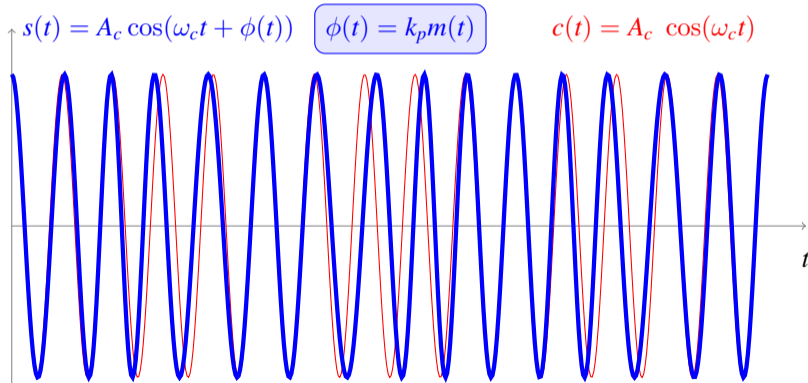
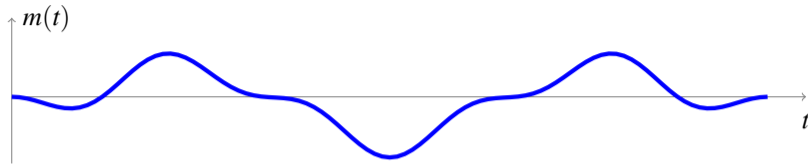
# Waveform of an FM Modulation



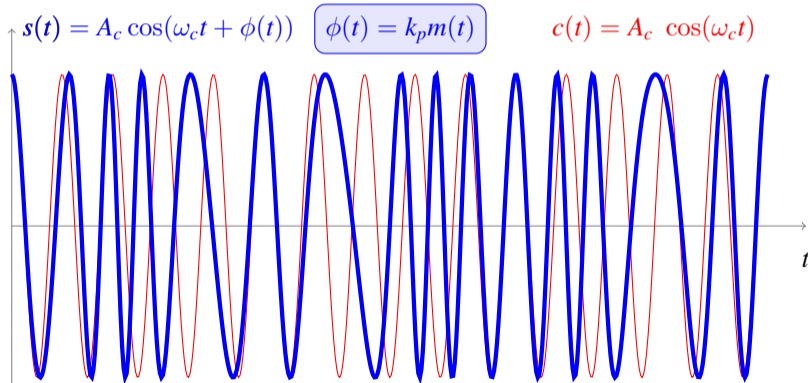
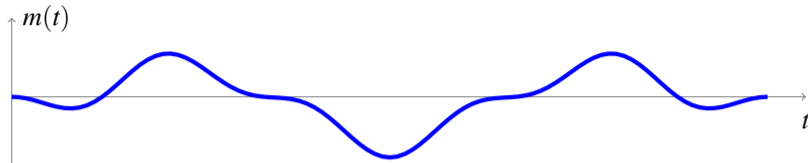
## Time offset: shift of a sinusoid (ahead or behind)



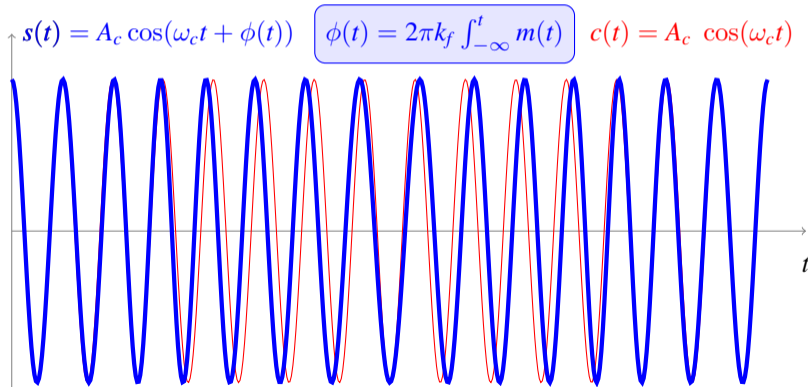
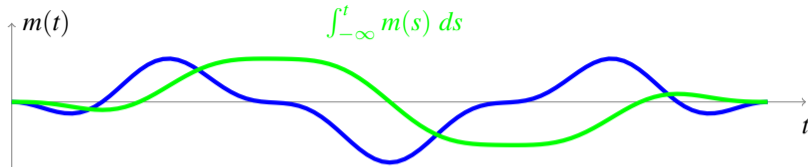
# PM Modulation - Signal for $k_p = 2\pi \times \frac{1}{4}$



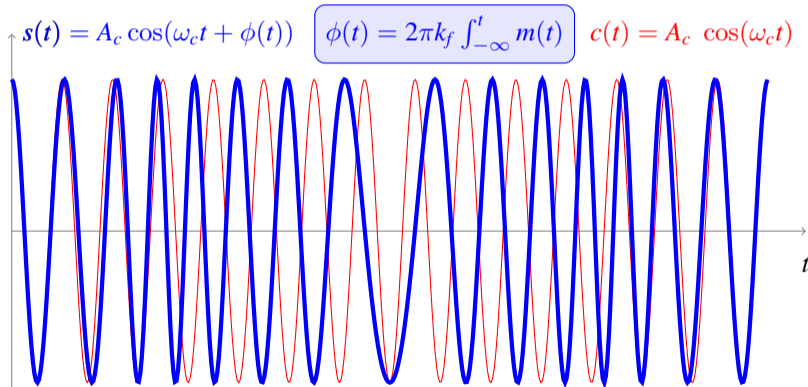
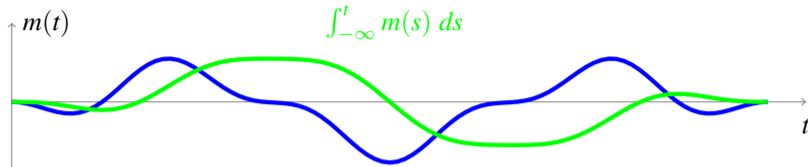
# PM Modulation - Signal for $k_p = 2\pi \times \frac{3}{4}$



# FM Modulation - Signal for $k_f = 2\pi \times \frac{1}{4}$



# FM Modulation - Signal for $k_f = 2\pi \times \frac{3}{4}$



## PM / FM relationship

- Phase modulation:

$$\phi(t) = k_p m(t)$$

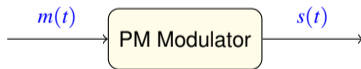
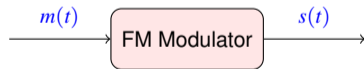
- Frequency modulation:

$$\Delta f_i(t) = \frac{1}{2\pi} \frac{d}{dt} \phi(t) = k_f m(t)$$

- Expressions of  $\phi(t)$  and  $\frac{d}{dt} \phi(t)$  in PM and FM

$$\phi(t) = \begin{cases} k_p m(t) & \text{PM} \\ 2\pi k_f \int_{-\infty}^t m(\tau) d\tau & \text{FM} \end{cases}$$

$$\frac{d}{dt} \phi(t) = \begin{cases} k_p \frac{d}{dt} m(t) & \text{PM} \\ 2\pi k_f m(t) & \text{FM} \end{cases}$$





## Modulation index $\beta$

- It defines the main characteristics of an angle modulation
  - ▶ Bandwidth
  - ▶ Signal to noise ratio (SNR)
- Maximum deviations
  - ▶ PM: maximum phase deviation

$$\Delta\phi_{\max} = k_p \max(|m(t)|) = k_p C_M$$

- ▶ FM: maximum frequency deviation

$$\Delta f_{\max} = k_f \max(|m(t)|) = k_f C_M$$

- Modulation indexes of PM and FM modulations

$$\beta_p = \Delta\phi_{\max} = k_p \max(|m(t)|) = k_p C_M$$

$$\beta_f = \frac{\Delta f_{\max}}{B} = \frac{k_f \max(|m(t)|)}{B} = \frac{k_f C_M}{B}$$

- ▶  $B$ : bandwidth in Hz of the modulating signal  $m(t)$

## Spectral analysis of angle modulations

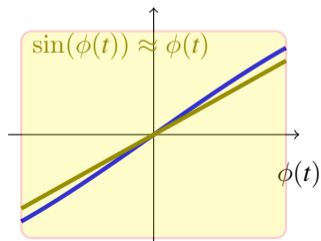
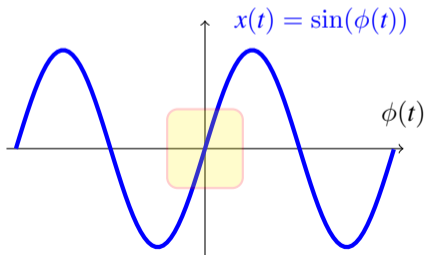
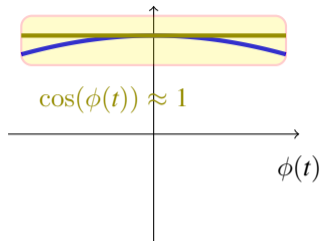
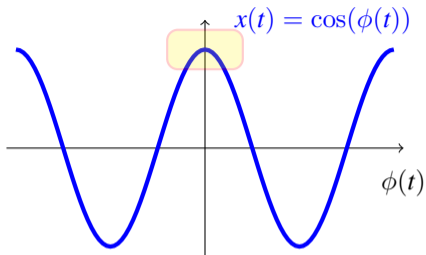
- A general analysis is difficult
  - ▶ Non-linear relationship between  $m(t)$  and  $s(t)$

$$s(t) = A_c \cos(\omega_c t + \phi(t))$$

$$\phi(t) = \begin{cases} k_p m(t) & \text{PM} \\ 2\pi k_f \int_{-\infty}^t m(\tau) d\tau & \text{FM} \end{cases}$$

- Calculation (exact or approximate) of the spectral representation
  - ▶ Narrowband angle modulations
  - ▶ Angle modulations with sinusoidal modulating signal
  - ▶ Angle modulations with periodic modulating signal
- Heuristic rule for calculating bandwidth
  - ▶ Angle modulations with generic modulating signal
    - ★ Carson's rule: approximation of the bandwidth

# Approximations of cosine and sine (small argument)



## Narrow band angle modulation

- Narrowband angle modulation:  $\phi(t) \ll 1$ 
  - ▶ Constants  $k_p$  or  $k_f$  are small ( $\beta$  is small)
- Generic expression of the modulated signal

$$s(t) = A_c \cos(\omega_c t + \phi(t))$$

- ▶ Trigonometric relation:

$$\cos(A \pm B) = \cos(A) \cos(B) \mp \sin(A) \sin(B)$$

- ▶ Approximations considered (for small  $\phi(t)$ )

$$\cos(\phi(t)) \approx 1, \sin(\phi(t)) \approx \phi(t)$$

- Alternative expression for the modulated signal

$$\begin{aligned} s(t) &= A_c \cos(\omega_c t) \cos \phi(t) - A_c \sin(\omega_c t) \sin \phi(t) \\ &\approx A_c \cos(\omega_c t) - A_c \phi(t) \sin(\omega_c t) \end{aligned}$$

- ▶ Expression similar to that of a conventional AM modulation

$$\text{AM: } s(t) = A_c \cos(\omega_c t) + A_c m_a(t) \cos(\omega_c t)$$

$$\text{Remember that } \phi(t) = \begin{cases} k_p m(t) & \text{PM} \\ 2\pi k_f \int_{-\infty}^t m(\tau) d\tau & \text{FM} \end{cases}$$

## Narrow band angle modulation - Analysis

- Approximate expression for the modulated signal

$$s(t) \approx A_c \cos(\omega_c t) - A_c \phi(t) \sin(\omega_c t)$$

- ▶ Expression similar to that of a conventional AM modulation

$$s(t) = A_c \cos(\omega_c t) + A_c m_a(t) \cos(\omega_c t)$$

- Spectrum of the modulated signal (considering the approximation)

- ▶ Two deltas, located at  $\pm\omega_c$  (carrier spectrum)
- ▶ Replicas of the spectrum of  $\phi(t)$  located at  $\omega = \pm\omega_c$
- ▶ Spectrum shape of  $\phi(t)$ 
  - ★ PM: proportional to the spectrum of  $m(t)$

$$\phi(t) = k_p m(t) \leftrightarrow \Phi(j\omega) = k_p M(j\omega)$$

- ★ FM: proportional to the spectrum of the integral of  $m(t)$

$$\phi(t) = 2\pi k_f \int_{-\infty}^t m(\tau) d\tau \leftrightarrow \Phi(j\omega) = 2\pi k_f \frac{M(j\omega)}{j\omega}$$

- Bandwidth (similar to conventional AM)

$$W_{NB} \approx 2 W \text{ rad/s}, \quad B_{NB} \approx 2 B \text{ Hz}$$

## Narrow band angle modulation - Summary

- Signal in time domain

$$s(t) \approx A_c \cos(\omega_c t) - A_c \phi(t) \sin(\omega_c t)$$

$$\phi(t) = \begin{cases} k_p m(t) & \text{PM} \\ 2\pi k_f \int_{-\infty}^t m(\tau) d\tau & \text{FM} \end{cases}$$

- Fourier transform

$$S(j\omega) \approx A_c \pi [\delta(\omega - \omega_c) + \delta(\omega + \omega_c)] - \frac{A_c}{2j} [\Phi(j\omega - j\omega_c) - \Phi(j\omega + j\omega_c)]$$

$$\Phi(j\omega) = \begin{cases} k_p M(j\omega) & \text{PM} \\ 2\pi k_f \frac{M(j\omega)}{j\omega} & \text{FM} \end{cases}$$

- Power spectral density

$$S_S(j\omega) \approx \frac{A_c^2}{2} \pi [\delta(\omega - \omega_c) + \delta(\omega + \omega_c)] + \frac{A_c^2}{4} [S_\Phi(j\omega - j\omega_c) + S_\Phi(j\omega + j\omega_c)]$$

$$S_\Phi(j\omega) = \begin{cases} k_p^2 S_M(j\omega) & \text{PM} \\ (2\pi k_f)^2 \frac{S_M(j\omega)}{\omega^2} & \text{FM} \end{cases}$$

## Modulation by a sinusoidal signal

- Sinusoidal modulating signal of amplitude  $a$  and frequency  $\omega_m$  rad/s

$$m(t) = \begin{cases} a \sin(\omega_m t) & \text{PM} \\ a \cos(\omega_m t) & \text{FM} \end{cases}$$

- Modulation indices of a PM and FM modulation

$$\beta_p = \Delta\phi_{\max} = k_p \max(|m(t)|) = k_p C_M = k_p a$$

$$\beta_f = \frac{\Delta f_{\max}}{B} = \frac{k_f \max(|m(t)|)}{B} = \frac{k_f C_M}{B} = k_f a \frac{2\pi}{\omega_m}$$

- Expressions of the phase term  $\phi(t)$

- ▶ Expressions of  $\phi(t)$  for PM

$$\phi(t) = k_p m(t) = k_p a \sin(\omega_m t) = \beta_p \sin(\omega_m t)$$

- ▶ Expressions of  $\phi(t)$  for FM

$$\phi(t) = 2\pi k_f \int_{-\infty}^t m(\tau) d\tau = 2\pi k_f a \frac{1}{\omega_m} \sin(\omega_m t) = \beta_f \sin(\omega_m t)$$

- Modulated signal: common expression for PM and FM

$$s(t) = A_c \cos(\omega_c t + \phi(t)) = A_c \cos(\omega_c t + \beta \sin(\omega_m t))$$

## Modulation by a sinusoidal signal (II)

- Sinusoidal modulating signal

$$m(t) = \begin{cases} a \sin(\omega_m t) & \text{PM} \\ a \cos(\omega_m t) & \text{FM} \end{cases}$$

- Modulated signal

$$s(t) = A_c \cos(\omega_c t + \beta \sin(\omega_m t)) = \operatorname{Re} \left( A_c e^{j\omega_c t} e^{j\beta \sin(\omega_m t)} \right)$$

The function  $e^{j\beta \sin(\omega_m t)}$  is periodic with frequency  $f_m = \frac{\omega_m}{2\pi}$  Hz

$$\text{Fourier series expansion: } e^{j\beta \sin(\omega_m t)} = \sum_{n=-\infty}^{\infty} J_n(\beta) e^{j(n \omega_m)t}$$

Index coefficient  $n$  of the series expansion:  $J_n(\beta)$

$J_n(\beta)$ : Bessel function of the first kind of order  $n$  and argument  $\beta$

- Alternative expression of the modulated signal

$$s(t) = \operatorname{Re} \left( A_c e^{j\omega_c t} \sum_{n=-\infty}^{\infty} J_n(\beta) e^{j(n \omega_m)t} \right) = \operatorname{Re} \left( \sum_{n=-\infty}^{\infty} A_c J_n(\beta) \underbrace{e^{j\omega_c t} e^{j(n \omega_m)t}}_{e^{j(\omega_c + n \omega_m)t}} \right)$$

$$s(t) = \sum_{n=-\infty}^{\infty} A_c J_n(\beta) \cos((\omega_c + n \omega_m) t)$$



## Modulation by a sinusoidal signal - Analysis

- The modulated signal contains sinusoids with the frequencies

$$\text{Frequencies (Hz) : } f_c + n f_m, \quad \text{for } n = 0, \pm 1, \pm 2, \dots$$

$$\text{Ang. Freq. (rad/s) : } \omega_c + n \omega_m, \quad \text{for } n = 0, \pm 1, \pm 2, \dots$$

$$\text{Amplitudes : } A_c J_n(\beta)$$

- Effective Bandwidth (contains 98% of the power):

$$B_e = 2 (\beta + 1) f_m \text{ Hz}$$

$$B_e = 2 (\beta + 1) f_m = \begin{cases} 2(k_p a + 1) f_m & \text{PM} \\ 2 \left( \frac{k_f a}{f_m} + 1 \right) f_m & \text{FM} \end{cases} = \begin{cases} 2(k_p a + 1) f_m & \text{PM} \\ 2(k_f a + f_m) & \text{FM} \end{cases}$$

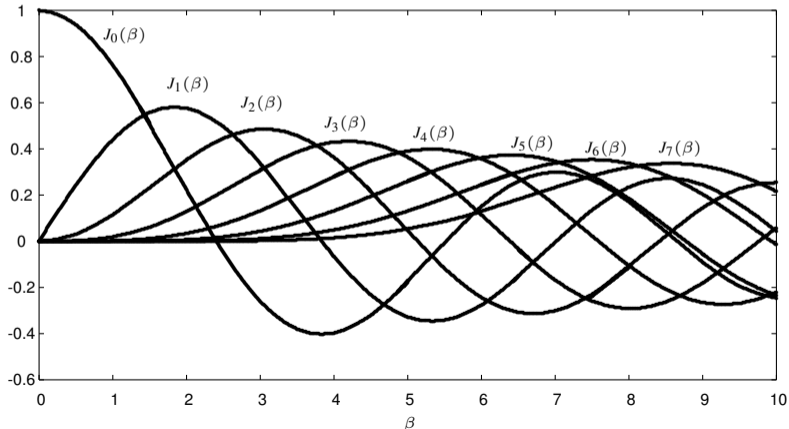
- Total number of harmonics in the effective bandwidth  $B_e$

$$M_e = 2 \lfloor \beta \rfloor + 3 = \begin{cases} 2 \lfloor k_p a \rfloor + 3 & \text{PM} \\ 2 \left\lfloor \frac{k_f a}{f_m} \right\rfloor + 3 & \text{FM} \end{cases}$$

# Bessel functions $J_n(\beta)$

$$J_n(\beta) = \sum_{k=0}^{\infty} \frac{(-1)^k \left(\frac{\beta}{2}\right)^{n+2k}}{k!(k+n)!}, \text{ For } \beta \downarrow J_n(\beta) \approx \frac{\beta^n}{2^n n!}, J_{-n}(\beta) = \begin{cases} J_n(\beta), & n \text{ even} \\ -J_n(\beta), & n \text{ odd} \end{cases}$$

It is usually found in tables or figures



## Bessel functions $J_n(\beta)$

$n$	$\beta = 0.1$	$\beta = 0.2$	$\beta = 0.5$	$\beta = 1$	$\beta = 2$	$\beta = 5$	$\beta = 8$	$\beta = 10$
0	0.9975	0.9900	0.9385	0.7652	0.2239	-0.1776	0.1717	-0.2459
1	0.0499	0.0995	0.2423	0.4401	0.5767	-0.3276	0.2346	0.0435
2	0.0012	0.0050	0.0306	0.1149	0.3528	0.0466	-0.1130	0.2546
3		0.0002	0.0026	0.0196	0.1289	0.3648	-0.2911	0.0584
4			0.0002	0.0025	0.0340	0.3912	-0.1054	-0.2196
5				0.0002	0.0070	0.2611	0.1858	-0.2341
6					0.0012	0.1310	0.3376	-0.0145
7					0.0002	0.0534	0.3206	0.2167
8						0.0184	0.2235	0.3179
9						0.0055	0.1263	0.2919
10						0.0015	0.0608	0.2075
11						0.0004	0.0256	0.1231
12						0.0001	0.0096	0.0634
13							0.0033	0.0290
14							0.0010	0.0120
15							0.0003	0.0045
16							0.0001	0.0016

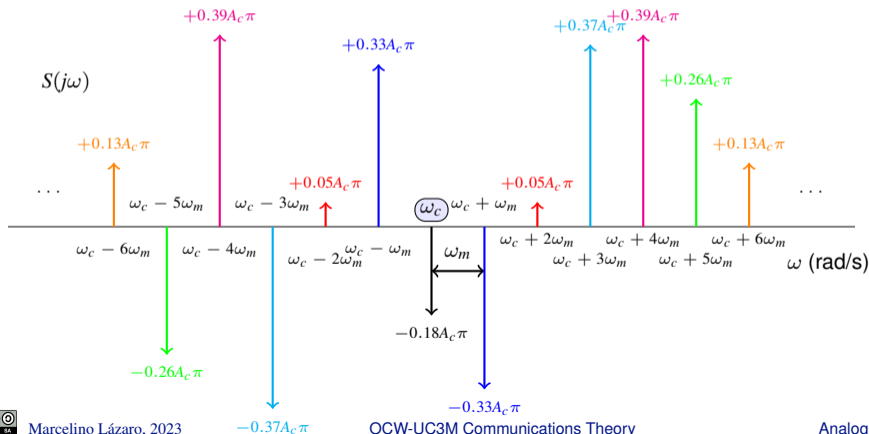
## Spectrum Shape - Example - Modulation with $\beta = 5$

- Sum of sinusoids of amplitude  $A_c J_n(\beta)$  and frequencies  $\omega_c + n \omega_m$  rad/s

► Spectrum: sum of deltas

- ★ Frequencies:  $\omega_c + n \omega_m$
- ★ Amplitudes:  $A_c J_n(\beta) \pi$
- ★ Number of harmonics in effective bandwidth:  $M_e = 2\lfloor\beta\rfloor + 3 = 13$

$$J_0(5) = -0.18, J_1(5) = -0.33, J_2(5) = 0.05, J_3(5) = 0.37, J_4(5) = 0.39, J_5(5) = 0.26, J_6(5) = 0.13 \dots$$



## Other types of modulators

- Modulation by means of a periodic signal
  - ▶ Supports a Fourier series expansion
    - ★ Sum of sinusoids of multiple frequencies of the one that defines the period
      - Frequencies in the spectrum of the signal

$$f_c \pm n f_m \text{ or } (\omega_c \pm n \omega_m)$$

- Amplitudes of each frequency: sum of the contributions of each harmonic

- Modulation by means of a non-periodic deterministic signal
  - ▶ Complicated analysis due to non-linearity
  - ▶ The [Carson's Rule](#) (heuristic rule) is applied

For modulating signals with bandwidth  $B$  Hz

$$B_{Carson} \approx 2 (\beta + 1) B \text{ Hz}$$

Bandwidth dependent on the modulation index  $\beta$

# NOISE EFFECT

## AT

# ANALOG MODULATIONS

# Effect of noise on amplitude modulations

## ● Premises

### ▶ Modulating signal $m(t)$

- ★ Bandwidth  $B$  Hz
- ★ Power  $P_M$  Watts

### ▶ Received signal

- ★ Transmission on Gaussian channel  
Ideal transmission without attenuation, without distortion, only with thermal noise

$$r(t) = s(t) + n(t)$$

- ★  $P_S$  : Power of the signal term (at the input of the receiver)

### ▶ Thermal noise: usual statistical model

- ★ Random process  $n(t)$  stationary, ergodic, white, Gaussian, with power spectral density  $S_n(j\omega) = \frac{N_0}{2}$

### ▶ Amplitude modulations: Coherent receiver

- ★ Filters will be introduced to limit the effect of noise
- ★ The filters will be considered as ideal (limit performance)

- The signal-to-noise ratio of the demodulated signal is analyzed for the different types of modulation and will be compared with the signal-to-noise ratio of the signal in a baseband transmission  $\left(\frac{S}{N}\right)_b$

## Reference - Baseband Transmission

- The unmodulated signal is transmitted:  $s(t) = m(t) \rightarrow P_S = P_M$
- Signal at receiver:  $r(t) = s(t) + n(t)$ 
  - ▶ There is usually attenuation during transmission
    - ★ In this analysis it is omitted for simplicity
    - ★ It is trivial to include it in development:  $s(t) \rightarrow \alpha s(t), P_S \rightarrow \alpha^2 P_S$
- Filtering in the receiver to minimize the effect of noise
  - ▶ Ideal low-pass filter of bandwidth  $B$  Hz



- ▶ Noise power at the filter output

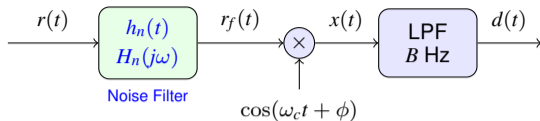
$$P_{n_f} = \frac{1}{2\pi} \int_{-\infty}^{+\infty} S_{n_f}(j\omega) d\omega = \frac{1}{2\pi} \int_{-2\pi B}^{+2\pi B} \frac{N_0}{2} d\omega = N_0 B \text{ Watts}$$

- Baseband signal-to-noise ratio

$$\left(\frac{S}{N}\right)_b = \frac{P_S}{N_0 B}, \quad \left(\frac{S}{N}\right)_b \text{ (dB)} = 10 \log_{10} \frac{P_S}{N_0 B} \text{ dB}$$



## Coherent Receiver + Noise Filtering - Notation



- Noise filter  $h_n(t) \overset{\mathcal{F}\mathcal{T}}{\leftrightarrow} H_n(j\omega)$

- ▶ Ideal band-pass filter: depends on the modulation that is used
  - ★ Passband and bandwidth: same as modulated signal  $s(t)$

- Coherent Receiver:  $\phi = \phi_c$  (for simplicity,  $\phi_c = 0$ )

- Received signal:  $r(t) = s(n) + n(t)$

- Filtered signal - filter suitable for modulation:  $s(t) * h_n(t) = s(t)$

$$r_f(t) = s(n) + n_f(t), \text{ with } n_f(t) = n(t) * h_n(t)$$

- Demodulated signal

$$x(t) = r_f(t) \times \cos(\omega_c t) = s(t) \cos(\omega_c t) + n_f(t) \cos(\omega_c t) = x_S(t) + x_n(t)$$

- Filtered demodulated signal

$$d(t) = x(t) * h_{LPF}(t) = x_S(t) * h_{LPF}(t) + x_n(t) * h_{LPF}(t) = d_S(t) + d_n(t)$$

# Coherent Receiver + Noise Filtering - Analysis

- Receiver output: Signal term  $d_S(t)$ 
  - ▶ Not affected by noise filter
  - ▶ For amplitude modulations, it was calculated previously

Modulation	$P_S$ (Watts)	$d_S(t)$	$P_{d_S}$ (Watts)
Conventional AM	$\frac{A_c^2}{2} [1 + P_{M_a}]$	$\frac{A_c}{2} [1 + m_a(t)]$	$\frac{A_c^2}{4} P_{M_a}$
DSB	$\frac{A_c^2}{2} P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$
SSB	$A_c^2 P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$
VSB	$A_c^2 P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$

$P_S$  : power of the modulated signal at the input of the receiver

$P_{d_S}$ : power in  $d_S(t)$  relative to  $m(t)$  -  $P_{M_a}$ : power of  $m_a(t)$ ,  $P_{M_a} = \frac{a^2}{C_M} P_M$

- Receiver output: Noise term  $d_n(t)$ 
  - ▶ Depends on the noise filter used
    - ★ Depends on the type of modulation
    - ★ Power:  $P_{d_n}$

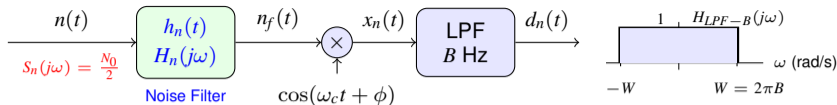
- Signal-to-noise ratio after demodulation:

$$\left(\frac{S}{N}\right)_d = \frac{P_{d_S}}{P_{d_n}}$$

- ▶ It will be compared with the baseband signal-to-noise ratio:

$$\left(\frac{S}{N}\right)_b = \frac{P_S}{N_0 B}$$

# Noise power after demodulation - General



- Power spectral density of the filtered noise  $n_f(t)$

$$S_{n_f}(j\omega) = S_n(j\omega) |H_n(j\omega)|^2 = \frac{N_0}{2} |H_n(j\omega)|^2$$

- Power spectral density of the demodulated noise  $x_n(t)$

$$S_{x_n}(j\omega) = \frac{1}{4} S_{n_f}(j\omega - j\omega_c) + \frac{1}{4} S_{n_f}(j\omega + j\omega_c) = \frac{N_0}{8} \left[ |H_n(j\omega - j\omega_c)|^2 + |H_n(j\omega + j\omega_c)|^2 \right]$$

- Power spectral density after low-pass filtering  $d_n(t)$

$$S_{d_n}(j\omega) = S_{x_n}(j\omega) |H_{LPF-B}(j\omega)|^2 = \begin{cases} S_{x_n}(j\omega), & \text{if } |\omega| \leq W = 2\pi B \\ 0, & \text{if } |\omega| > W = 2\pi B \end{cases}$$

- Power after low pass filtering

$$\begin{aligned} P_{d_n} &= \frac{1}{2\pi} \int_{-\infty}^{\infty} S_{d_n}(j\omega) d\omega = \frac{1}{2\pi} \int_{-2\pi B}^{2\pi B} S_{x_n}(j\omega) d\omega \\ &= \frac{N_0}{8} \left[ \frac{1}{2\pi} \int_{-2\pi B}^{2\pi B} |H_n(j\omega - j\omega_c)|^2 d\omega + \frac{1}{2\pi} \int_{-2\pi B}^{2\pi B} |H_n(j\omega + j\omega_c)|^2 d\omega \right] \end{aligned}$$

## Noise power after demodulation - General (II)

The noise power at the output of a synchronous AM receiver is

$$P_{d_n} = \frac{N_0}{8} \times F_0 \text{ Watts} \quad \text{with} \quad F_0 = \frac{1}{2\pi} \int_{-W}^W |H_n(j\omega - j\omega_c)|^2 d\omega + \frac{1}{2\pi} \int_{-W}^W |H_n(j\omega + j\omega_c)|^2 d\omega$$

Making simple changes of variables ( $\omega' = \omega - \omega_c$  and  $\omega' = \omega + \omega_c$ )

$$F_0 = \frac{1}{2\pi} \int_{-\omega_c - W}^{-\omega_c + W} |H_n(j\omega')|^2 d\omega' + \frac{1}{2\pi} \int_{\omega_c - W}^{\omega_c + W} |H_n(j\omega')|^2 d\omega'$$

Considering the even symmetry of  $|H_n(j\omega)|$

$$F_0 = 2 \times F, \quad \text{with} \quad F = \frac{1}{2\pi} \int_{\omega_c - W}^{\omega_c + W} |H_n(j\omega)|^2 d\omega$$

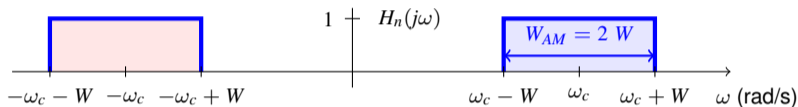
$$P_{d_n} = \frac{N_0}{4} \times F$$

## Calculation of noise power - conventional AM and DSB

- For both modulations the noise filter is identical

$$H_n(j\omega) = \begin{cases} 1, & \text{if } \omega_c - W \leq |\omega| \leq \omega_c + W \\ 0, & \text{otherwise} \end{cases}$$

$W$ : bandwidth in rad/s ( $W = 2\pi B$ )

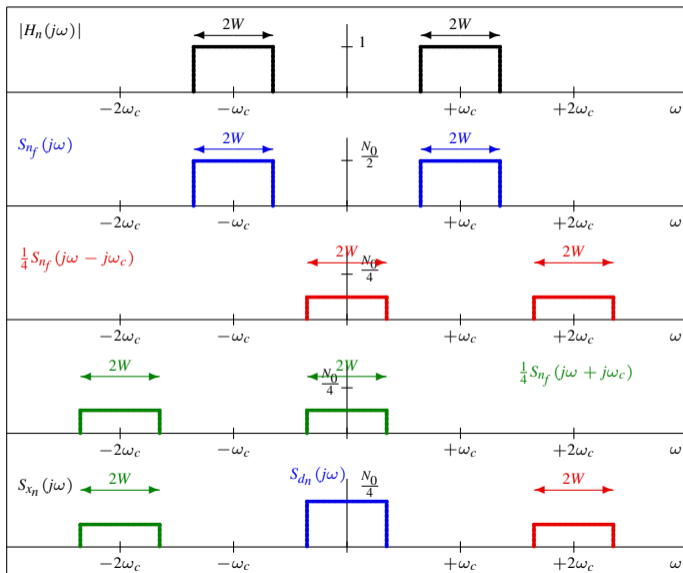


- Calculation of noise power

$$\begin{aligned} F &= \frac{1}{2\pi} \int_{\omega_c - W}^{\omega_c + W} |H_n(j\omega)|^2 d\omega = \frac{1}{2\pi} \int_{\omega_c - W}^{\omega_c + W} 1 d\omega \\ &= \frac{1}{2\pi} 2W = 2B \end{aligned}$$

$$P_{d_n} = \frac{N_0}{4} \times F = \frac{1}{2} N_0 B \text{ Watts}$$

# Noise in conventional AM and DSB - Frequency interpretation

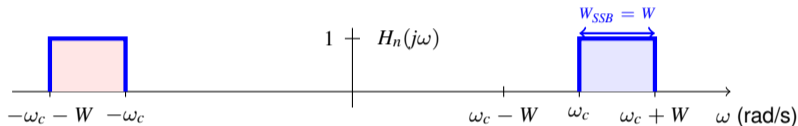


## Calculation of noise power - SSB

- The case of upper sideband is presented

$$H_n(j\omega) = \begin{cases} 1, & \text{if } \omega_c \leq |\omega| \leq \omega_c + W \\ 0, & \text{otherwise} \end{cases}$$

$W$ : bandwidth in rad/s ( $W = 2\pi B$ )

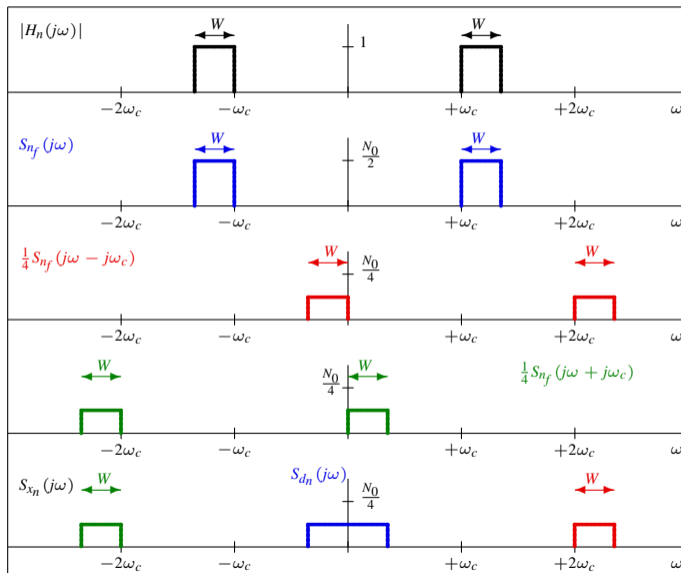


- Calculation of noise power

$$\begin{aligned} F &= \frac{1}{2\pi} \int_{\omega_c - W}^{\omega_c + W} |H_n(j\omega)|^2 d\omega = \frac{1}{2\pi} \int_{\omega_c}^{\omega_c + W} 1 d\omega \\ &= \frac{1}{2\pi} W = B \end{aligned}$$

$$P_{d_n} = \frac{N_0}{4} \times F = \frac{1}{4} N_0 B \text{ Watts}$$

# SSB Noise - Frequency Interpretation (USB)





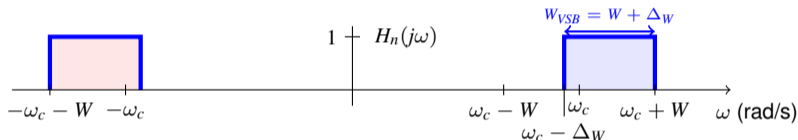
# Calculation of noise power - VSB

- The case of upper sideband is presented

$$H_n(j\omega) = \begin{cases} 1, & \text{if } \omega_c - \Delta_W \leq |\omega| \leq \omega_c + W \\ 0, & \text{otherwise} \end{cases}$$

$W$ : bandwidth in rad/s ( $W = 2\pi B$ )

$\Delta_W$ : bandwidth excess (vestige) in rad/s ( $\Delta_W = 2\pi\Delta_B$ )

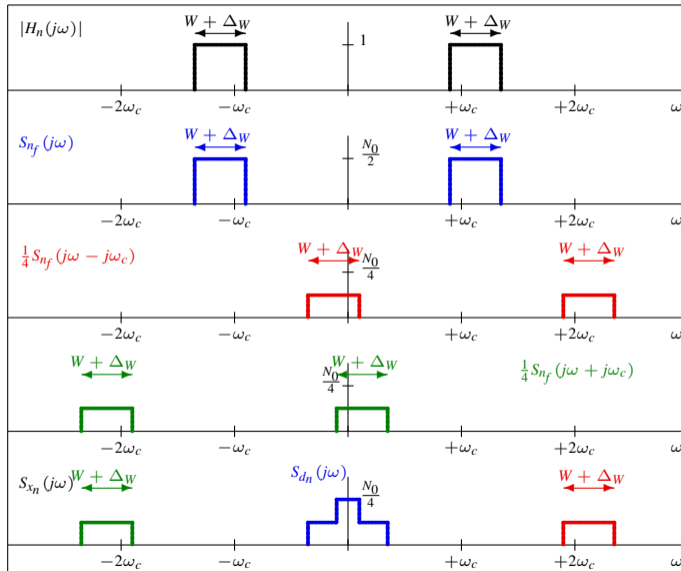


- Calculation of noise power

$$\begin{aligned} F &= \frac{1}{2\pi} \int_{\omega_c - W}^{\omega_c + W} |H_n(j\omega)|^2 d\omega = \frac{1}{2\pi} \int_{\omega_c - \Delta_W}^{\omega_c + W} 1 d\omega \\ &= \frac{1}{2\pi} (W + \Delta_W) = (B + \Delta_B) \end{aligned}$$

$$P_{dn} = \frac{N_0}{4} \times F = \frac{1}{4} N_0 (B + \Delta_B) \text{ Watts}$$

# Noise in VSB - Frequency Interpretation (USB)



## Signal to noise ratio at the output of a coherent receiver

Modulation	$P_S$	$d_S(t)$	$P_{d_S}$	$P_{d_n}$
Conventional AM	$\frac{A_c^2}{2} [1 + P_{M_a}]$	$\frac{A_c}{2} [1 + m_a(t)]$	$\frac{A_c^2}{4} P_{M_a}$	$\frac{1}{2} N_0 B$
DSB	$\frac{A_c^2}{2} P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$	$\frac{1}{2} N_0 B$
SSB	$A_c^2 P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$	$\frac{1}{4} N_0 B$
VSB	$A_c^2 P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$	$\frac{1}{4} N_0 (B + \Delta_B)$

$P_S$ : signal power at the input of the receiver    Assumption:  $r(t) = s(t) + n(t)$

- Signal to noise ratio

$$\left( \frac{S}{N} \right) = \frac{P_{d_S}}{P_{d_n}}$$

- Signal to noise ratio compared with a baseband transmission

$$\left( \frac{S}{N} \right) = \eta \left( \frac{S}{N} \right)_b = \eta \frac{P_S}{N_0 B}$$

Analysis of the result:  $\eta \begin{matrix} \geq \\ \leq \end{matrix} 1$

## Calculation of signal to noise ratios - DSB and SSB

Modulation	$P_S$	$d_S(t)$	$P_{d_S}$	$P_{d_n}$
Conventional AM	$\frac{A_c^2}{2} [1 + P_{M_a}]$	$\frac{A_c}{2} [1 + m_a(t)]$	$\frac{A_c^2}{4} P_{M_a}$	$\frac{1}{2} N_0 B$
DSB	$\frac{A_c^2}{2} P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$	$\frac{1}{2} N_0 B$
SSB	$A_c^2 P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$	$\frac{1}{4} N_0 B$
VSB	$A_c^2 P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$	$\frac{1}{4} N_0 (B + \Delta_B)$

- Signal to noise ratio for DSB

$$\left(\frac{S}{N}\right)_{DSB} = \frac{P_{d_S}}{P_{d_n}} = \frac{\frac{A_c^2}{4} P_M}{\frac{1}{2} N_0 B} = \frac{\frac{A_c^2}{2} P_M}{N_0 B} = \frac{P_S}{N_0 B} = \left(\frac{S}{N}\right)_b$$

Same signal to noise ratio as transmitting in baseband

- Signal to noise ratio for SSB

$$\left(\frac{S}{N}\right)_{SSB} = \frac{P_{d_S}}{P_{d_n}} = \frac{\frac{A_c^2}{4} P_M}{\frac{1}{4} N_0 B} = \frac{A_c^2 P_M}{N_0 B} = \frac{P_S}{N_0 B} = \left(\frac{S}{N}\right)_b$$

Same signal to noise ratio as transmitting in baseband

# Calculation of signal to noise ratios - Conventional AM

Modulation	$P_S$	$d_S(t)$	$P_{d_S}$	$P_{d_n}$
Conventional AM	$\frac{A_c^2}{2} [1 + P_{M_a}]$	$\frac{A_c}{2} [1 + m_a(t)]$	$\frac{A_c^2}{4} P_{M_a}$	$\frac{1}{2} N_0 B$
DSB	$\frac{A_c^2}{2} P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$	$\frac{1}{2} N_0 B$
SSB	$A_c^2 P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$	$\frac{1}{4} N_0 B$
VSB	$A_c^2 P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$	$\frac{1}{4} N_0 (B + \Delta_B)$

- Signal to noise ratio for conventional AM

$$\begin{aligned} \left(\frac{S}{N}\right)_{AM} &= \frac{P_{d_S}}{P_{d_n}} = \frac{\frac{A_c^2}{4} P_{M_a}}{\frac{1}{2} N_0 B} = \frac{\frac{A_c^2}{2} P_{M_a}}{N_0 B} = \frac{P_{M_a}}{1 + P_{M_a}} \frac{\frac{A_c^2}{2} [1 + P_{M_a}]}{N_0 B} \\ &= \underbrace{\frac{P_{M_a}}{1 + P_{M_a}}}_{\eta_{AM}} \frac{P_S}{N_0 B} = \eta_{AM} \left(\frac{S}{N}\right)_b \end{aligned}$$

- Worse than baseband transmission: Efficiency factor  $\eta_{AM} < 1$

$$\eta_{AM} = \frac{P_{M_a}}{1 + P_{M_a}} = \frac{\frac{a^2}{C_M^2} P_M}{1 + \frac{a^2}{C_M^2} P_M} = \frac{P_M}{\frac{C_M^2}{a^2} + P_M}$$

Depends on the modulation index: worse efficiency for lower values of  $a$

## Calculation of signal to noise ratios - VSB

Modulation	$P_S$	$d_S(t)$	$P_{d_S}$	$P_{d_n}$
Conventional AM	$\frac{A_c^2}{2} [1 + P_{Ma}]$	$\frac{A_c}{2} [1 + m_a(t)]$	$\frac{A_c^2}{4} P_{Ma}$	$\frac{1}{2} N_0 B$
DSB	$\frac{A_c^2}{2} P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$	$\frac{1}{2} N_0 B$
SSB	$A_c^2 P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$	$\frac{1}{4} N_0 B$
VSB	$A_c^2 P_M$	$\frac{A_c}{2} m(t)$	$\frac{A_c^2}{4} P_M$	$\frac{1}{4} N_0 (B + \Delta_B)$

- Signal to noise ratio for VSB

$$\begin{aligned} \left(\frac{S}{N}\right)_{VSB} &= \frac{P_{d_S}}{P_{d_n}} = \frac{\frac{A_c^2}{4} P_M}{\frac{1}{4} N_0 (B + \Delta_B)} = \frac{A_c^2 P_M}{N_0 (B + \Delta_B)} = \frac{B}{B + \Delta_B} \frac{A_c^2 P_M}{N_0 B} \\ &= \underbrace{\frac{B}{B + \Delta_B}}_{\eta_{VSB}} \frac{P_S}{N_0 B} = \eta_{VSB} \left(\frac{S}{N}\right)_b \end{aligned}$$

- ▶ **Worse than baseband transmission:** Efficiency factor  $\eta_{VSB} < 1$

$$\eta_{VSB} = \frac{B}{B + \Delta_B}$$

Depends on bandwidth excess  $\Delta_B$  (vestige): usually  $\Delta_B \ll B$  and in that case  $\eta_{VSB} \approx 1$ , i.e., signal-to-noise ratio is relatively close to  $(S/N)_b$

## Noise in angle modulations

- Much more complex analysis (non-linear dependency)
- Demodulator output (noisy)

$$d(t) = \begin{cases} k_p m(t) + Y_n(t), & \text{PM} \\ k_f m(t) + \frac{1}{2\pi} \frac{d}{dt} Y_n(t), & \text{FM} \end{cases}$$

$Y_n(t)$ : noise term at the demodulator output

- Signal to noise ratio (for received signal power  $P_S = \frac{A_c^2}{2}$ )

$$\left(\frac{S}{N}\right)_d = \begin{cases} \frac{k_p^2 A_c^2}{2} \frac{P_M}{N_0 B} = P_M \left(\frac{\beta_p}{\max |m(t)|}\right)^2 \left(\frac{S}{N}\right)_b, & \text{PM} \\ \frac{3k_f^2 A_c^2}{2B^2} \frac{P_M}{N_0 B} = 3P_M \left(\frac{\beta_f}{\max |m(t)|}\right)^2 \left(\frac{S}{N}\right)_b, & \text{FM} \end{cases}$$

Signal-to-noise gain proportional to modulation index squared

- General expression

$$\left(\frac{S}{N}\right)_d = \alpha \left(\frac{P_M}{C_M^2}\right) \times \beta^2 \times \left(\frac{S}{N}\right)_b$$

- ▶ The factor  $\alpha$  depends on the modulation:  $\alpha_{PM} = 1, \alpha_{FM} = 3$
- ▶ The term  $\left(\frac{P_M}{C_M^2}\right)$  is usually constant (depends on the type of signals)

## Noise in angle modulations - Threshold effect

- The gain in  $\left(\frac{S}{N}\right)_d$  is only obtained from a threshold signal-to-noise ratio at the receiver input

$$\left(\frac{S}{N}\right)_{threshold} = 20 (\beta + 1)$$

- This implies in practice that there is a threshold level of received power that must be reached

$$P_{R_{threshold}} = (N_0 B) \times \left(\frac{S}{N}\right)_{threshold} \rightarrow A_{c,threshold} = \sqrt{2P_{R_{threshold}}}$$