



Analog modulations

Communication Theory

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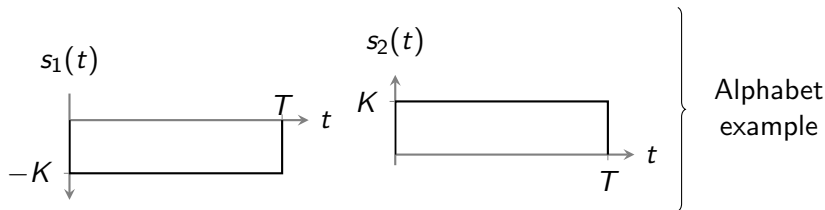
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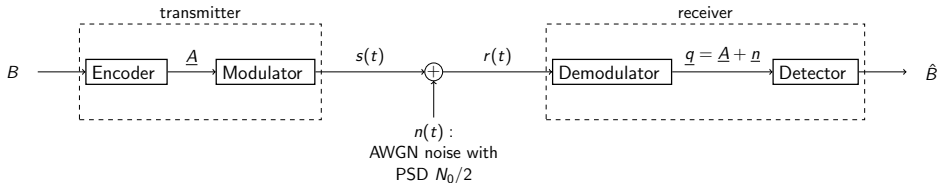
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Analog vs digital communications systems

In **digital** systems, information is carried by symbols in an alphabet,

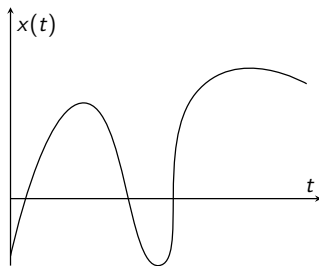


and the most basic scheme of a system is...



Analog vs digital communications systems

In **analog** systems, information is contained in a continuous waveform...



2 possibilities:

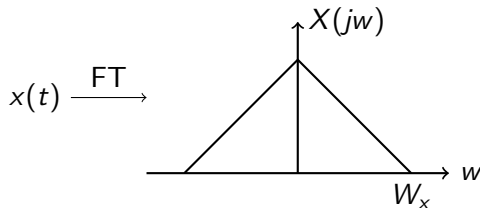
- discretize the signal and transmit it using a **digital** system
- transmit it directly using an **analog** system.

We focus on the latter approach...2 types of channels:

- **Baseband**
- **Passband**

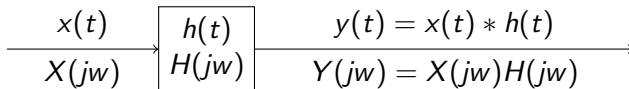
Transmission of a baseband signal

Assume we want to transmit



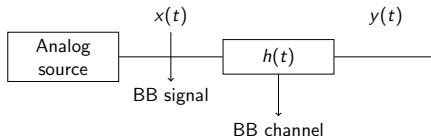
- baseband signal with
- bandwidth W_x

...and we model the channel as a **Linear Time Invariant** system...

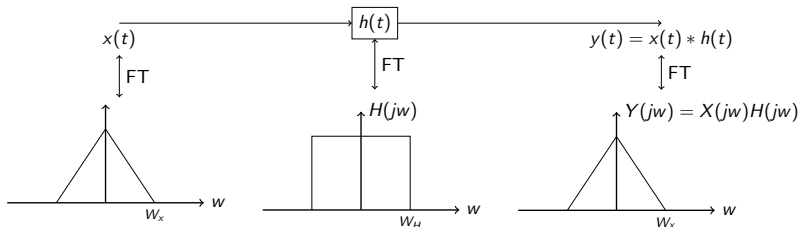


Baseband transmission

The information signal is transmitted as is



Assuming an ideal **baseband** channel whose bandwidth is larger than that of the signal ($W_H > W_x$)...

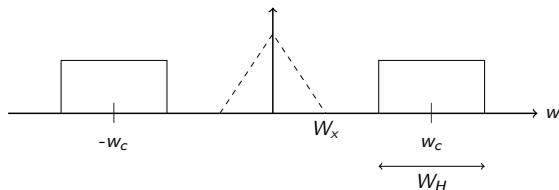


Examples: telephone subscriber loop, public address systems, closed-circuit TV

Passband transmission

Let us consider an ideal **passband** channel whose

- 1 bandwidth is larger than that of the signal, and whose
- 2 center frequency, w_c , is **much** larger than the bandwidth of the signal.



- **passband** channel with
- bandwidth W_H

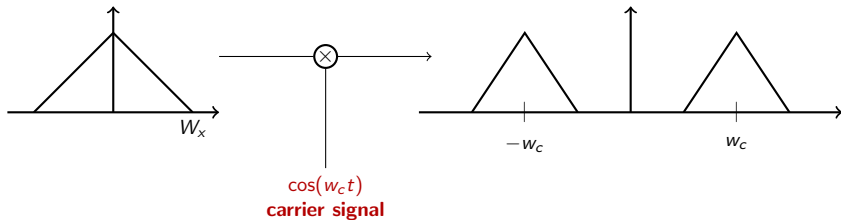
- 2 entails $w_c - \frac{W_H}{2} > W_x$, and hence

$$Y(jw) = X(jw)H(jw) = 0,$$

i.e., the signal **cannot** get through.

Modulation

This is solved by shifting the spectrum of $x(t)$...



Mathematically,

$$x(t) \cos(\omega_c t) \xleftrightarrow{FT} \frac{1}{2} X(j(\omega - \omega_c)) + \frac{1}{2} X(j(\omega + \omega_c))$$

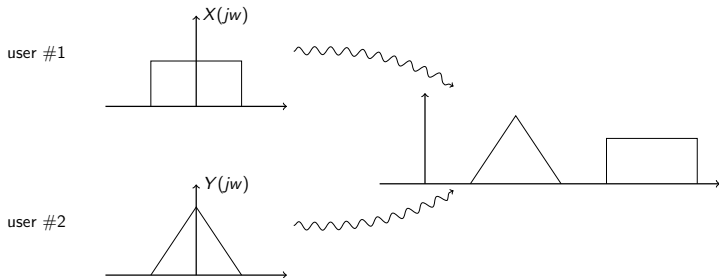
This operation is called **modulation**. It involves three signals

$$\underbrace{y(t)}_{\text{modulated signal}} = \underbrace{x(t)}_{\text{modulating signal}} \cdot \underbrace{\cos(\omega_c t)}_{\text{carrier signal}}$$

Modulation

Other uses:

- multiuser systems



- protection against
 - noise
 - unauthorized users listening in the channel

Types of modulation

Let us consider an information signal, $x(t)$, that is

- a **baseband** signal, i.e., $X(j\omega) = 0, \forall |\omega| > W$
 - a realization of a band-limited WSS random process $X(t)$ with $S_X(j\omega) = 0, \forall |\omega| > W$
- a *power signal*, i.e., whose power is finite

Transmission of $x(t)$ is achieved by embedding it in a **carrier** signal of the form

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

amplitude frequency phase

and $x(t)$ can be *modulating*

- the **amplitude**,
- the **frequency**, or
- the **phase**.

Linear vs. angular modulations

In general, the modulated signal has the form

$$y(t) = r(t) \cos(w_c t + \varphi(t))$$

- $x(t)$ is embedded in $r(t) \rightarrow$ **linear** or **amplitude modulation**¹

$$y(t) = r(t) \cos(w_c t + \varphi)$$

with *constant* frequency and phase.

- $x(t)$ is embedded in $\varphi(t) \rightarrow$ **angular modulation**

$$y(t) = A \cos(w_c t + \varphi(t))$$

with *constant* amplitude and frequency.

¹ $\nRightarrow r(t) = x(t)$...since some transformation might be applied

Linear modulations

$$\begin{aligned} y(t) &= r(t) \cos(w_c t + \varphi) = r(t) (\cos(w_c t) \cos \varphi - \sin(w_c t) \sin \varphi) \\ &= \underbrace{r(t) \cos \varphi}_{\substack{x_i(t) \\ \text{in-phase com-} \\ \text{ponent}}} \cos(w_c t) - \underbrace{r(t) \sin \varphi}_{\substack{x_q(t) \\ \text{quadrature} \\ \text{component}}} \sin(w_c t), \end{aligned}$$

$x_i(t)$ and $x_q(t)$ are rewritten in a more convenient form

- $x_i(t) = r(t) \cos \varphi = A_c + A_m x(t)$
- $x_q(t) = r(t) \sin \varphi = A_n \tilde{x}(t)$

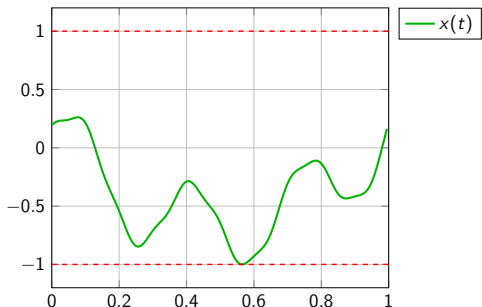
where $\tilde{x}(t)$ is a transformation of $x(t)$. The new parameters determine the type of modulation:

- $A_c \neq 0, A_m \neq 0, A_n = 0 \rightarrow$ **AM modulation** (*conventional amplitude modulation*)
- $A_c = 0, A_m \neq 0, A_n = 0 \rightarrow$ **DSB modulation** (Double Side Band modulation)
- $A_c = 0, A_m \neq 0, A_n \neq 0 \rightarrow$ **SSB modulation** (Single Side Band modulation)

AM modulation

$$y(t) = (A_c + A_m x(t)) \cos(w_c t)$$

Let us assume $|x(t)| \leq 1$ (if not, we can do $x_n(t) = \frac{x(t)}{\max|x(t)|}$,)



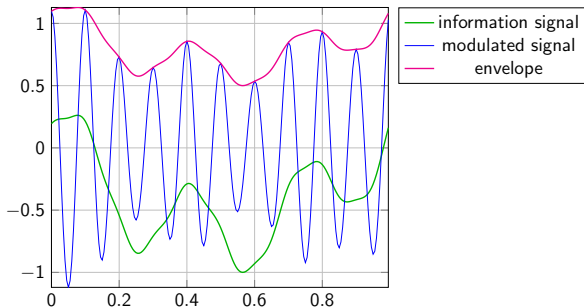
$$y(t) = \left(A_c + \frac{A_c}{A_c} A_m x(t) \right) \cos(w_c t) = A_c (1 + m x(t)) \cos(w_c t)$$

with

$$m = \frac{A_m}{A_c} \equiv \text{modulation index}$$

AM demodulation

$A_c (1 + mx(t)) \geq 0$ implies this signal matches the **envelope** of the **modulated signal**, $y(t)$



Envelope of a signal

The **envelope** of an oscillating signal (for example, a cosine) is a *smooth* signal that outlines its extremes.

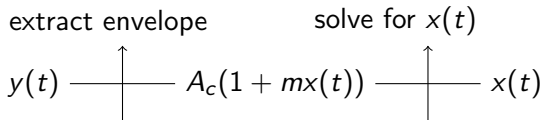
AM demodulation

In order to guarantee $A_c(1 + mx(t)) \geq 0$, and assuming $|x(t)| \leq 1$ (i.e. it is *normalized*), it is enough to choose

$$0 < m \leq 1.$$

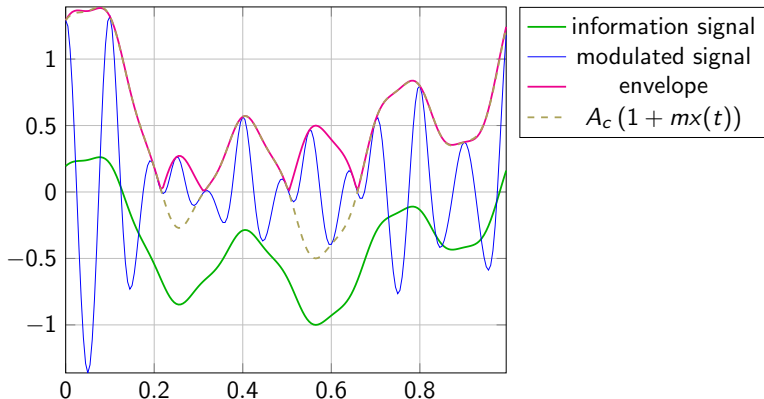
Indeed,

$$\left. \begin{array}{l} 0 < m \leq 1 \\ |x(t)| \leq 1 \end{array} \right\} \Rightarrow (1 + mx(t)) \geq 0 \Rightarrow \text{demodulation is very easy}$$



Overmodulation

If $m > 1$, it might happen that $A_c(1 + mx(t)) < 0$ at some point...



The envelope is **not** $A_c(1 + mx(t))$ anymore, and hence demodulation is not so easy. It is called **overmodulation**.

Angular modulations

The information is in the argument of the **carrier**

Drawbacks

- they are way **more complex** than linear modulations
 - sometimes, they need to be studied through approximations
- they take up **more bandwidth**

Advantages

- they are **less affected by noise...**
 - they trade off bandwidth for immunity against noise

Algebra for Angular modulations

$$y(t) = A \cos(w_c t + \varphi(t)) = A \cos \phi(t),$$

where

$$\phi(t) = w_c t + \varphi(t) \equiv \text{instantaneous phase}$$

From it, we have

$$\frac{d\phi(t)}{dt} = w_i(t) \frac{\text{rad}}{\text{second}} \equiv \text{instantaneous frequency},$$

in radians, which becomes

$$\frac{1}{2\pi} \frac{d\phi(t)}{dt} = f_i(t) \text{ Hz} \equiv \text{instantaneous frequency}.$$

in hertz.

Doing some algebra

$$f_i(t) = \frac{1}{2\pi} \frac{d\phi(t)}{dt} = \frac{1}{2\pi} \frac{d(w_c t + \varphi(t))}{dt} \stackrel{w_c=2\pi f_c}{=} f_c + \frac{1}{2\pi} \frac{d\varphi(t)}{dt}$$

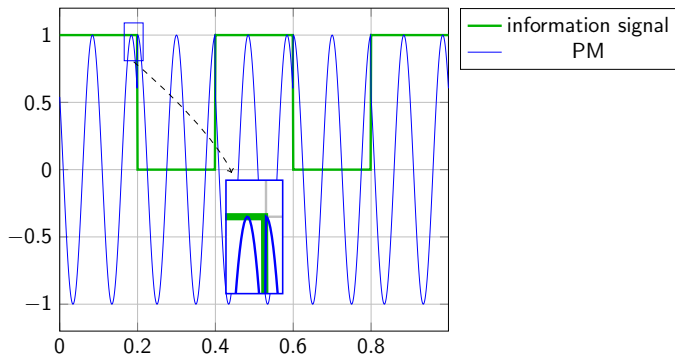
Phase modulation

Being $x(t)$ the information signal...

$$\varphi(t) = \beta x(t) \longrightarrow \text{Phase modulation (PM)}$$

with $\beta \equiv$ phase deviation constant. Hence,

$$y(t) = A \cos(\omega_c t + \beta x(t))$$



Frequency modulation

$$f_i(t) = f_c + f_d x(t) \longrightarrow \text{Frequency modulation (FM)}$$

with $f_d \equiv$ frequency deviation. It can be proved that

$$\varphi(t) = 2\pi f_d \int_{-\infty}^t x(u) du,$$

and hence

$$y(t) = A \cos \left(w_c t + 2\pi f_d \int_{-\infty}^t x(u) du \right)$$

