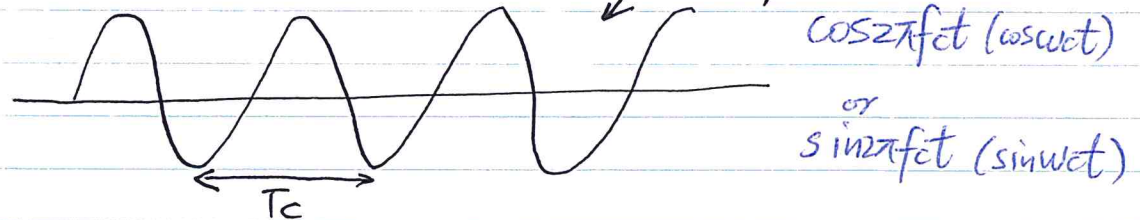


①

# ECES19B/496A

Basic concepts:

- ~~frequency band~~ carrier frequency



$$f_c = \frac{1}{T_c} \quad \omega_c = 2\pi f_c = \frac{2\pi}{T_c}$$

$f_c$ : several hundred MHz 1G-5G

1.9, 2.4 GHz

2G, 3G, 4G

sub-6 GHz

5G term

28GHz and above

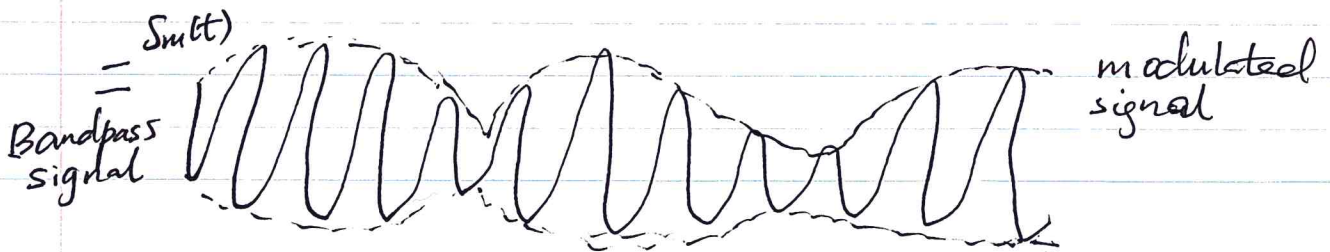
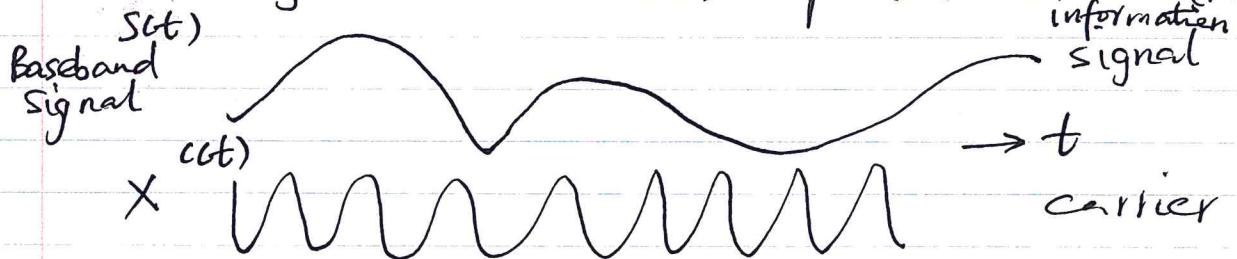
mmWave

- Channel bandwidth

\* carrier is just a sinusoid signal. Ideally its BW is 0.

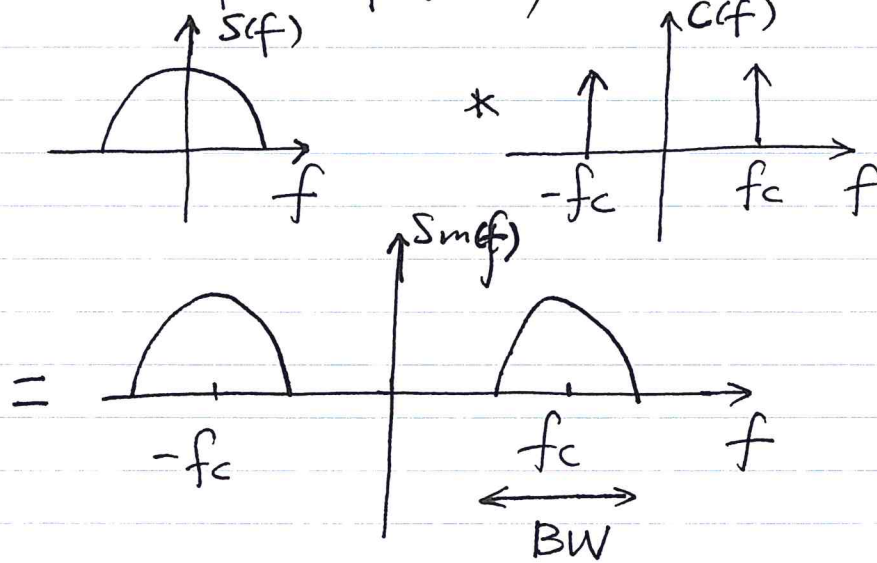
\* When the carrier signal is modulated to carry information data, the modulated signal has a bandwidth.

Ex1. Analog modulation: ~~SAM~~ Amplitude modulation (AM)



(2)

view from frequency domain

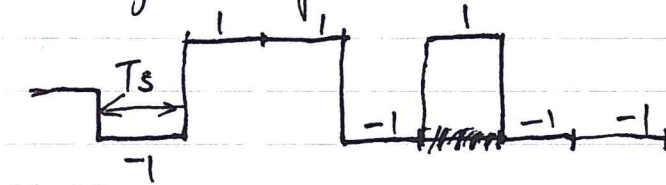


EX2. Digital modulation: pulse amplitude modulation

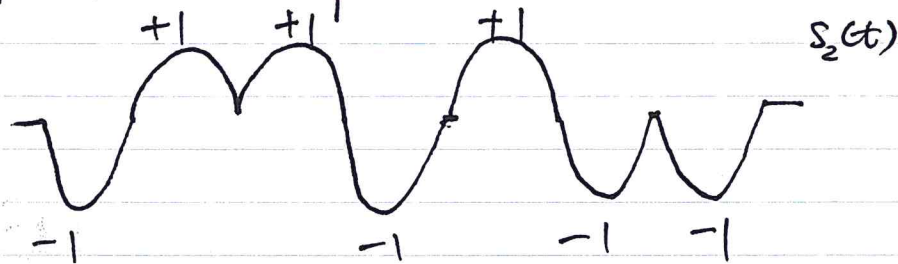
information data is binary digits (PAM)

0 1 1 0 1 0 0  $\xrightarrow{\text{bit mapping symbol}}$  -1, 1, 1, -1, 1, -1, -1

rectangular pulse



~~Half~~ Cosine pulse

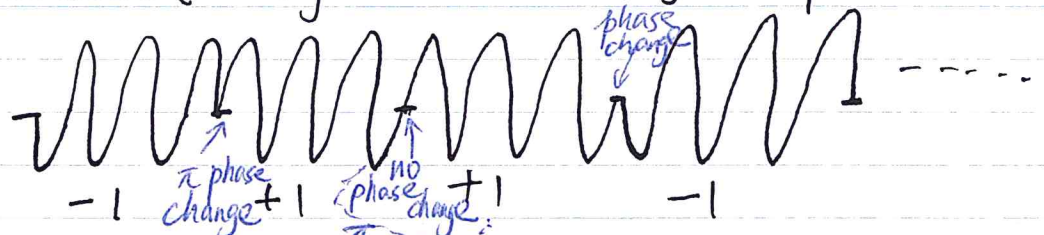


$S_1(t)$

$T_s$ : symbol duration

$R_s = \frac{1}{T_s}$ : symbol rate or Band rate

Modulated signal: rectangular pulse



$-\cos x = \cos(x + \pi)$  phase change

3

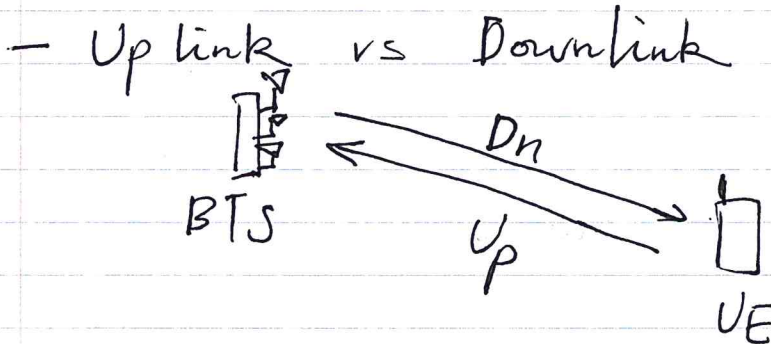
Modulated signal: half cosine pulse



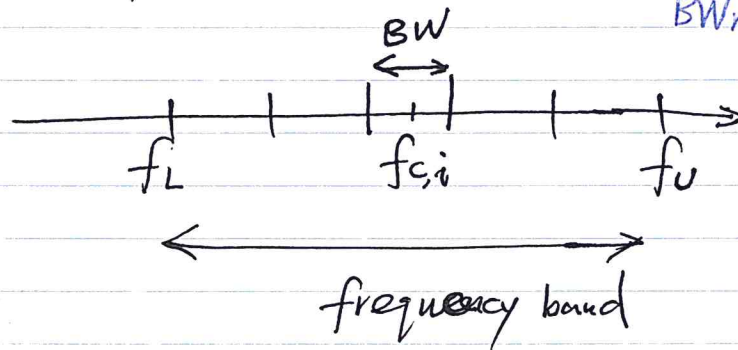
rect. pulse  
↓  
 $T_s$   
half cosine pulse

↓  
 $T_s$   
root raise cosine (RRC)  
↔  
 $4 \sim 6 T_s$

Digitally modulated signal's BW is largely determined by the BW of the shaping pulse.  
 $BW_{rect} > BW_{cosine} > BW_{RRC}$



— frequency band



$$\# \text{ channels} \approx \frac{f_u - f_L}{BW}$$

— TX power

Tx power determines the amplitude/magnitude of the modulated signal that is ~~radiated~~ fed by antenna for sent out.



④

$$20 \text{ dBm} = 10^{\frac{20}{10}} \text{ mW} = 10^2 \text{ mW} = 100 \text{ mW}$$

$$30 \text{ dBm} = 1 \text{ W}$$

$$-10 \text{ dBm} = 10^{\frac{-10}{10}} \text{ mW} = 0.1 \text{ mW}$$

$$0 \text{ dBm} = 10^0 \text{ mW} = 1 \text{ mW}$$

dB linear

$$X + 3 \text{ dB} = 2X$$

$$X + 10 \text{ dB} = 10X$$

— Data rate bps

Data rate is directly related to BW and transmission power. Will be discussed more later.

— Link budget

Link budget is related to the coverage distance that can be achieved with the received power above the minimum required for data detection.

## Communication principle Review

Take QPSK as example

Slides 30-31

At TX side,

0 1 1 0 ----

↓ ↓  
-d-jd d-jd

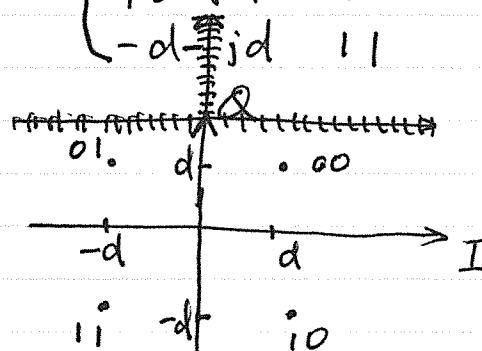
I: -d, d, ...

Q: d, -d, ...

QPSK

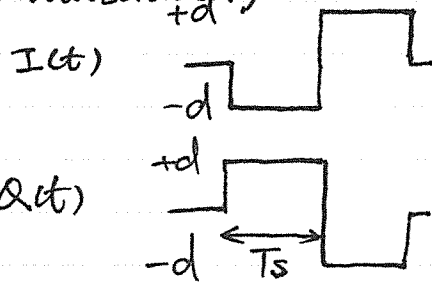
bit to symbol mapping rule

$$x_n = \begin{cases} d+jd & 00 \\ -d+jd & 01 \\ +d-jd & 10 \\ -d-jd & 11 \end{cases}$$



(5)

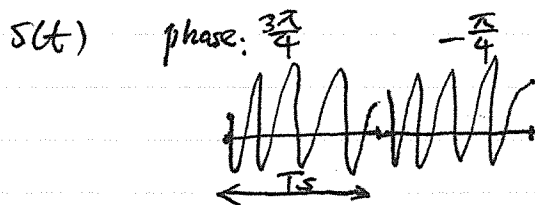
Let  $p(t)$  be a rectangular pulse (for ease of illustration)



$$s(t) = I(t)A \cos \omega_c t - Q(t)A \sin \omega_c t$$

1st symbol  $-dA \cos \omega_c t - dA \sin \omega_c t$   
 $= -dA (\cos \omega_c t + \sin \omega_c t) = -dA \sqrt{2} \cos(\omega_c t - \frac{\pi}{4})$   
 $= \sqrt{2} dA \cos(\omega_c t + \frac{3\pi}{4})$

2nd symbol  $dA \cos \omega_c t - (-d)A \sin \omega_c t$   
 $= dA (\cos \omega_c t + \sin \omega_c t)$  transmit power (average)  
 $= \sqrt{2} dA \cos(\omega_c t - \frac{\pi}{4})$   $P_{Tx} = \frac{(\sqrt{2} dA)^2}{2} = d^2 A^2$



$$E_b = \frac{(\sqrt{2} dA)^2}{2} \cdot \frac{1}{2} = \frac{d^2 A^2}{2} T$$

One  $T_s$  sends 2 bits

$$R_b = 2R_s = \frac{2}{T_s} \text{ bit rate}$$

$$T_b = \frac{T_s}{2} \text{ bit duration}$$

At RX side, linear convolution

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

If the channel is only additive white Gaussian noise (AWGN)  
 $h(t) = \delta(t)$

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

Assume lowpass filter (LPF) has coefficient of 1 for the passing band

1st symbol, for the I branch, ignoring noise

$$(r(t)A \cos \omega_c t)_{LPF} = (-d^2 A^2 \cos^2 \omega_c t - d^2 A^2 \sin^2 \omega_c t \cdot \cos \omega_c t)_{LPF}$$

⑥

$$= \left( -dA^2 \cdot \frac{1 + \cos 2\omega t}{2} - \frac{dA^2}{2} \sin 2\omega t \right) \Big|_{LPF}$$

$$= -\frac{dA^2}{2}$$

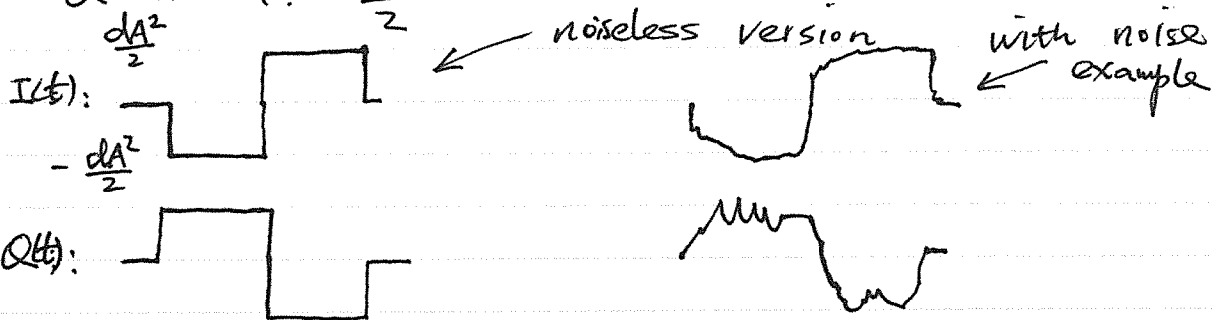
for the Q branch

$$(r(t) A \sin \omega t) \Big|_{LPF} = \frac{dA^2}{2} \leftarrow \frac{dA^2 \cdot \sin^2 \omega t}{2} \Big|_{LPF} = \frac{dA^2}{2} (1 - \cos 2\omega t) \Big|_{LPF}$$

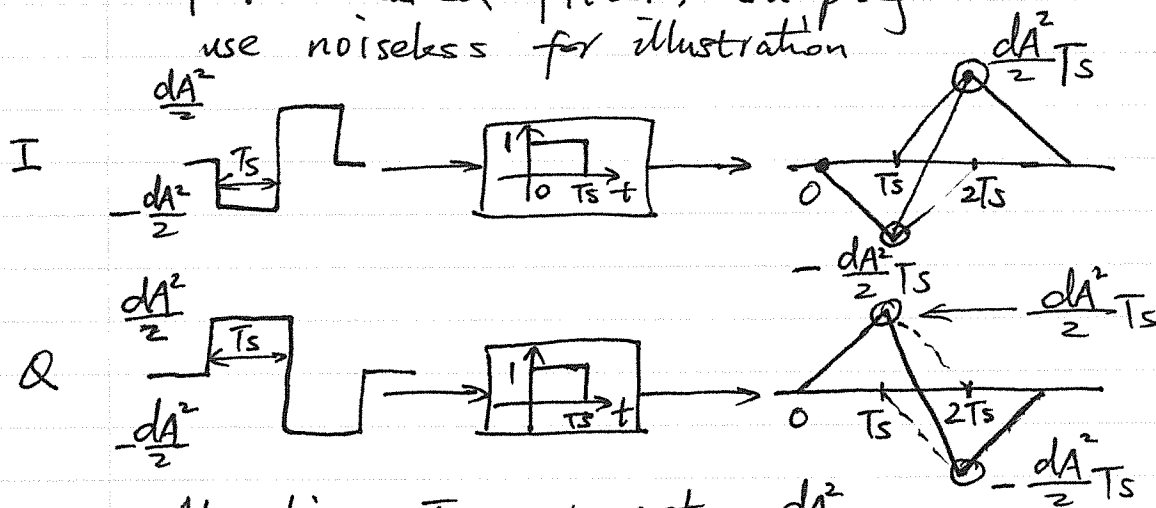
Similarly, 2nd symbol

$$I \text{ branch: } \frac{dA^2}{2}$$

$$Q \text{ branch: } -\frac{dA^2}{2}$$

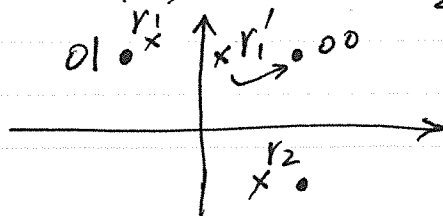


After matched filter, sampling  
use noiseless for illustration



At time  $T_s$ , we get  $-\frac{dA^2}{2} T_s$  for I and  $\frac{dA^2}{2} T_s$  for Q

...  $2T_s, \dots$   $\frac{dA^2}{2} T_s$  for I and  $-\frac{dA^2}{2} T_s$  for Q



$$r_1 = -\frac{dA^2}{2} T_s + j \frac{dA^2}{2} T_s \quad r_1' = r_1 + n_1$$

$$r_2 = \frac{dA^2}{2} T_s - j \frac{dA^2}{2} T_s \quad r_2' = r_2 + n_2$$

⑦

— With noise, <sup>if</sup> we actually get sample points  $r_1$  for 1st symbol at  $T_s$  and  $r_2$  at  $t = 2T_s$ .

$r_1$  is closest to signal point 01, so we determine (this is called maximum likelihood rule) ML rule  
01 was sent.

$r_2$  is closest to signal point 10, .. ..  
10 was sent.

— If we actually get sample points  $r_1'$ , because of large noise, 00 will be decided as the tx data, and then an error occurs.

∴ It is important how large the noise is. In other words, the signal-to-noise ratio (SNR) is a key benchmark to error performance.

In digital communications, SNR is defined as

$$SNR = \frac{E_b}{N_0}$$

$E_b$ : Average energy per bit  
 $N_0$ : Noise spectral density

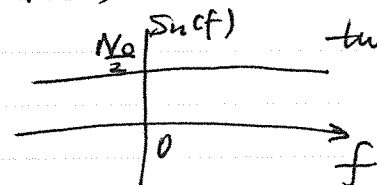
$n(t)$  is a random process with wide sense stationary property and zero mean, uncorrelated samples at two different time instants.

$$E[n(t)] = 0 \quad \text{white noise}$$

$$R_n(\tau) = E[n(t)n(t+\tau)] = \frac{N_0}{2} \delta(\tau) \quad \text{autocorrelation function}$$

power spectral density (PSD) is the Fourier transform of  $R_n(\tau)$ . two sided

$$S_n(f) = \frac{N_0}{2}, \quad -\infty < f < \infty$$



8

$N_0 = kT$      $k$ : Boltzmann's constant in Joules per kelvin  
 $T$ : temperature in kelvins.

$$k = 1.38 \times 10^{-23} \text{ J/K}$$

$$T = \cancel{288} \approx 288.4 \text{ K} \quad (15^\circ \text{C})$$

$$N_0 = 3.98 \times 10^{-21} \text{ J (W/Hz)}$$

$$= 3.98 \times 10^{-18} \text{ mW/Hz}$$

$$= -174 \text{ dBm/Hz}$$

noise power  $P_n = N_0 B$      $B$ : bandwidth  $\propto \frac{1}{T_s}$

$$\frac{E_b}{N_0} = \frac{E_b \cdot B}{N_0 \cdot B} \hookrightarrow \frac{E_b / T_s}{P_n} \propto \frac{P_{tx}}{P_n}$$

To model the received signal <sup>(discrete)</sup> after MF and sampling,

$$y = x_i + n$$

where  $E[x_i^2] = E_s$   $E_s$  is energy per symbol

$n = n_c + j n_s$ ,  $n_c$  and  $n_s$  are i.i.d. Gaussian

$$n_c \sim \mathcal{N}(0, \sigma_n^2), \quad n_s \sim \mathcal{N}(0, \sigma_n^2)$$

$$\sigma_n^2 = \frac{N_0}{2}$$

$$\text{or } n \sim \mathcal{CN}(0, N_0)$$

Often SNR is defined as

$$\text{SNR} = \frac{\text{average received signal energy per symbol time}}{\text{noise energy per symbol time}}$$

$$\text{So } E[x_i^2] = P$$

$$\text{SNR} = \frac{E[x_i^2]}{\sigma_n^2} = \frac{P}{N_0}$$