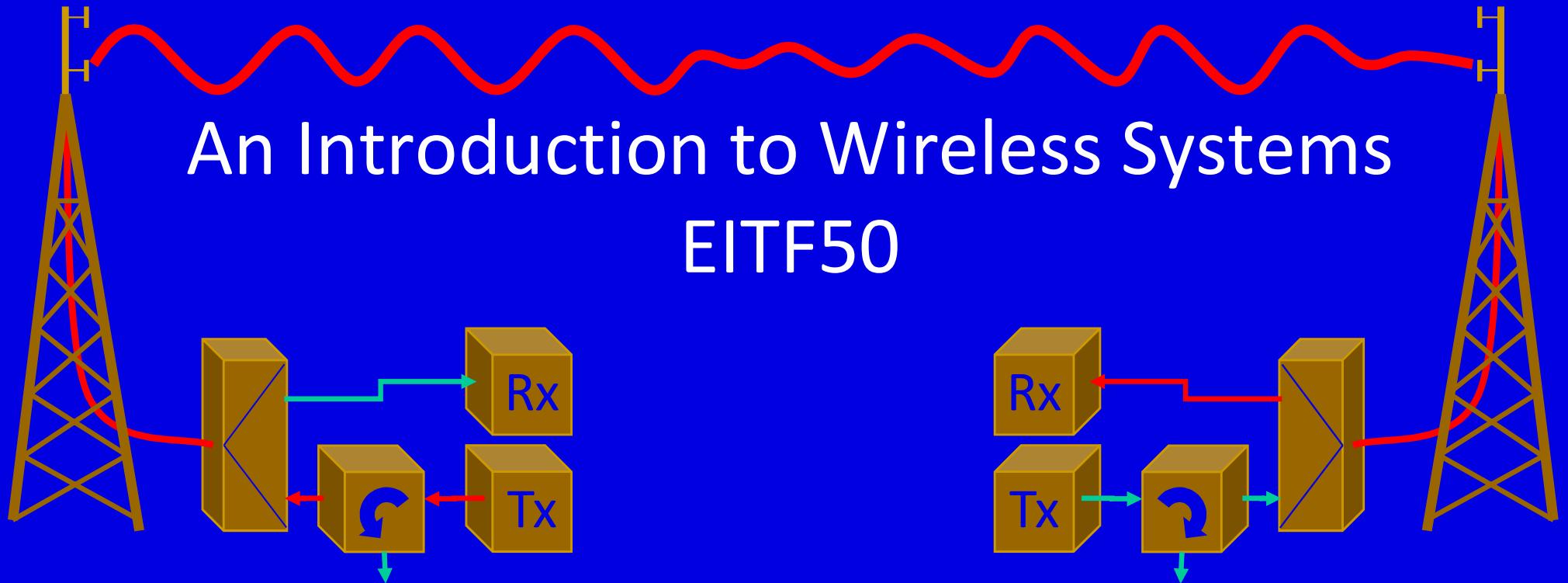


# Lecture 8

## An Introduction to Wireless Systems EITF50



Johan Wernehag  
Electrical and Information Technology  
Slides 21-34 by Ove Edfors



# Contents

- Digital Modulation
  - FSK, PSK and ASK
  - Noise and Bit Error Ratio (BER)
- Multi-User System (multiple access)
  - FDMA, TDMA, CDMA
- Spreading Technique
- Multi-carrier Technique
  - History and evolution
  - Transmitters and receivers

# Digital Modulation

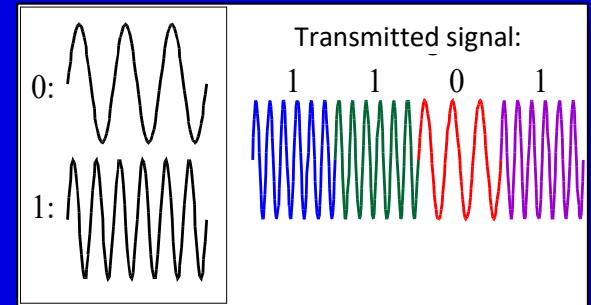
- A general radio signal may be written

$$s(t) = A \cos(\omega t + \theta)$$

there are three parameters to alternate: the amplitude  $A$ , frequency  $\omega$  and phase  $\theta$ , these leads to three basic forms of digital modulation:

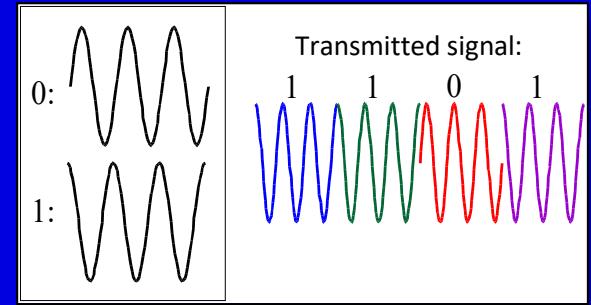
- FSK (Frequency Shift Keying)**  
The frequency carries the information.

Example:



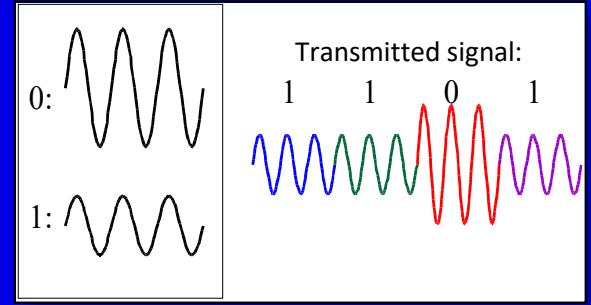
- PSK (Phase Shift Keying)**  
The phase carries the information.

Example:



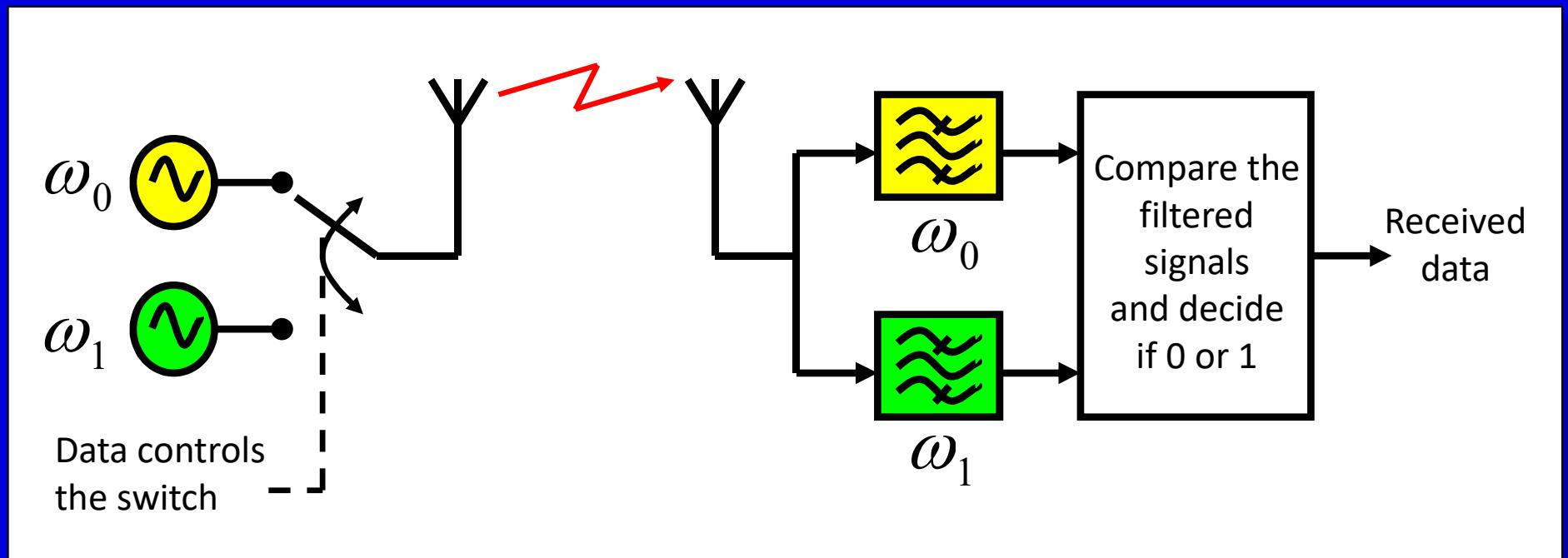
- ASK (Amplitude Shift Keying)**  
The amplitude carries the information.

Example:



# FSK - Frequency Shift Keying

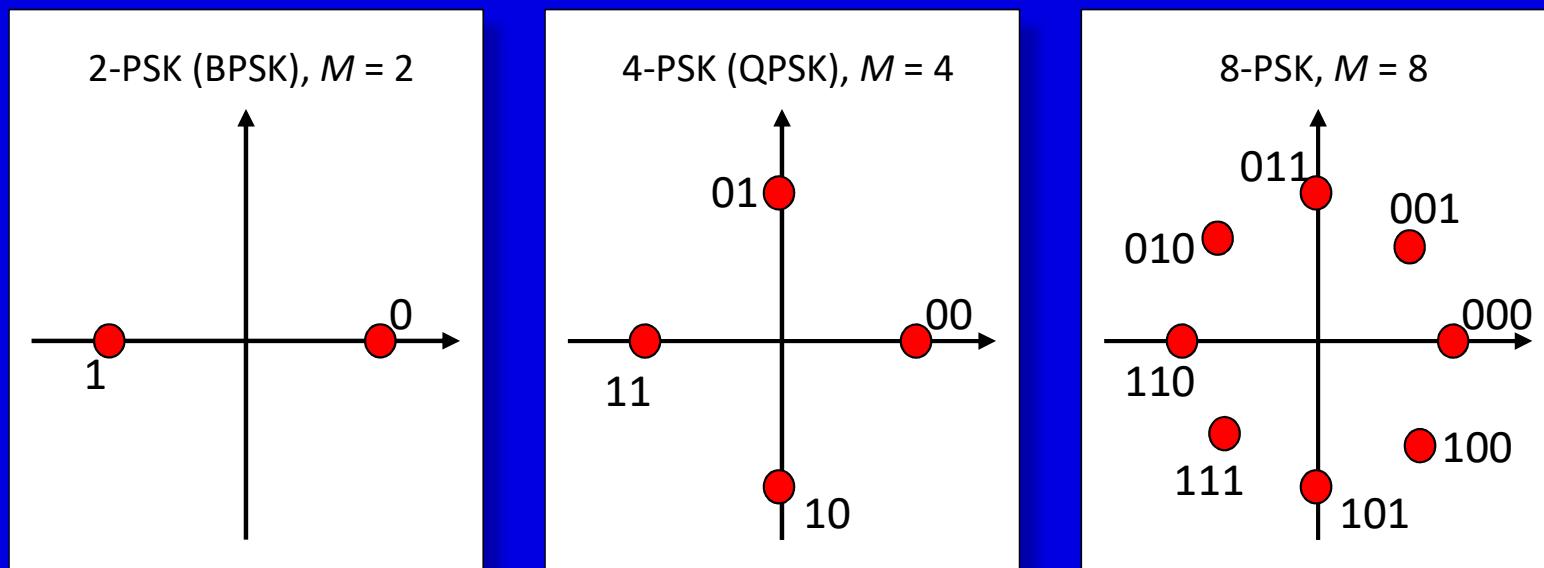
- At binary FSK 0 and 1 are represented by radio signals of different frequencies:



- Neither phase- nor amplitude references are needed in the receiver.
- Simple modulation and demodulation, but poor performance!

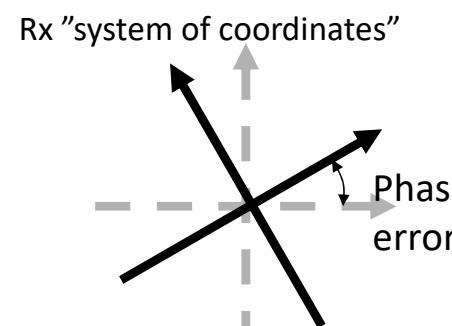
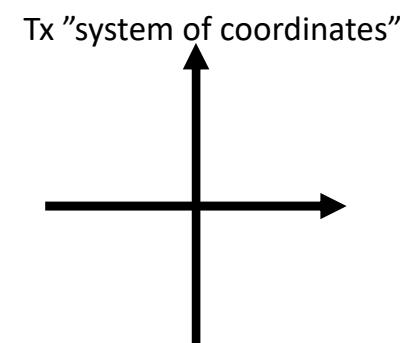
# PSK - Phase shift keying

- A PSK symbol with  $M = 2^n$  discrete phases may carry  $n$  information bits.



Does PSK show any problems?

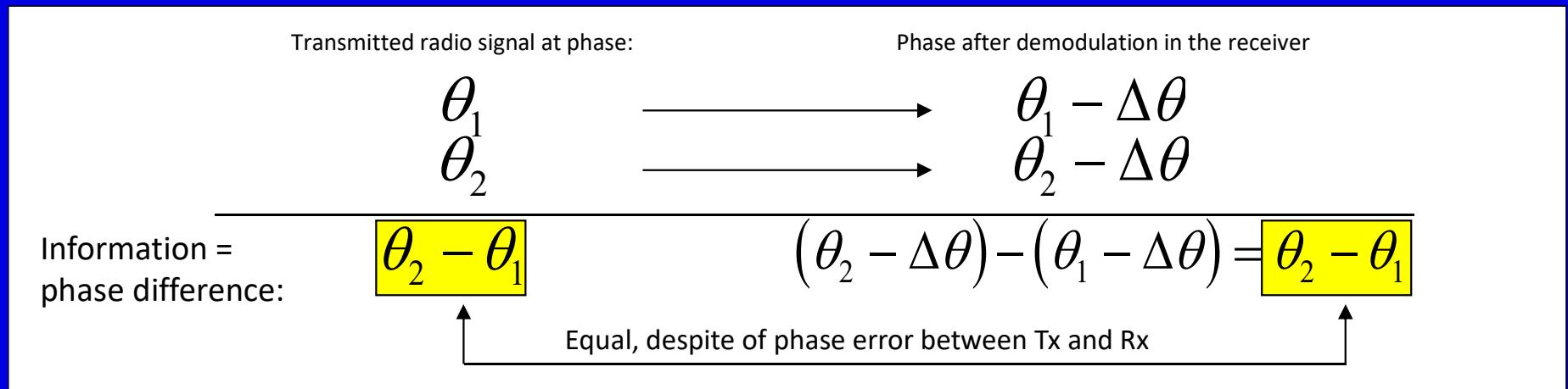
- Absolute phase reference is required in the receiver.



Can we find any "smart" cure?

# DPSK - Differential PSK

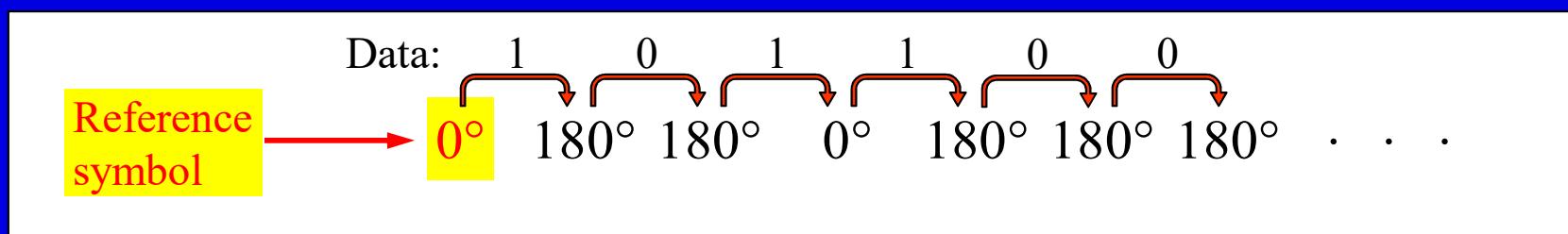
- Idea: Let the **phase difference** between two subsequent PSK symbols carry the information (instead of absolute phase).
- What is the effect of a phase error  $\Delta\theta$  between transmitter and receiver?
  - The phase of two subsequent PSK symbols are  $\theta_1$  and  $\theta_2$ .
  - The information is carried by the difference  $\theta_1 - \theta_2$ .



- Differential PSK (DPSK) eliminates the need for absolute phase reference!

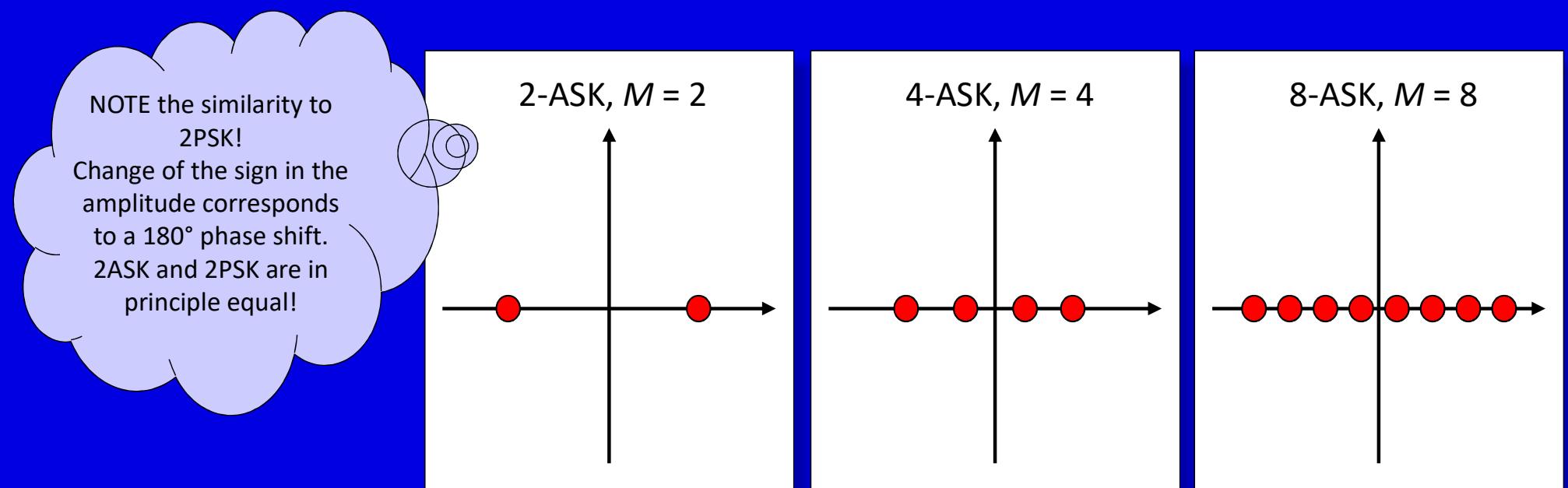
# Example: Differential BPSK (DBPSK)

- DPSK uses two phases,  $0^\circ$  and  $180^\circ$ .
- The transmission starts with a reference symbol at an arbitrary phase angle (for example  $0^\circ$ ).
- The binary information then controls the phase of the next symbol as:
  - If input data is 1, change the phase by  $180^\circ$ .
  - If input data is 0, keep the phase unchanged.
- The data sequence 101100... will be transmitted as BPSK symbols at phase angles as:



# ASK - Amplitude Shift Keying

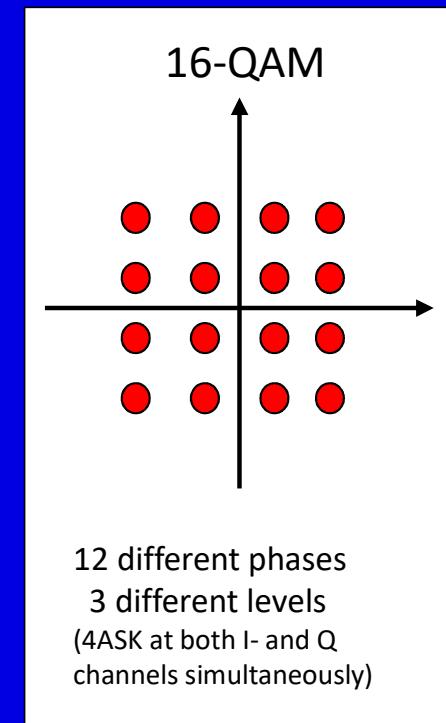
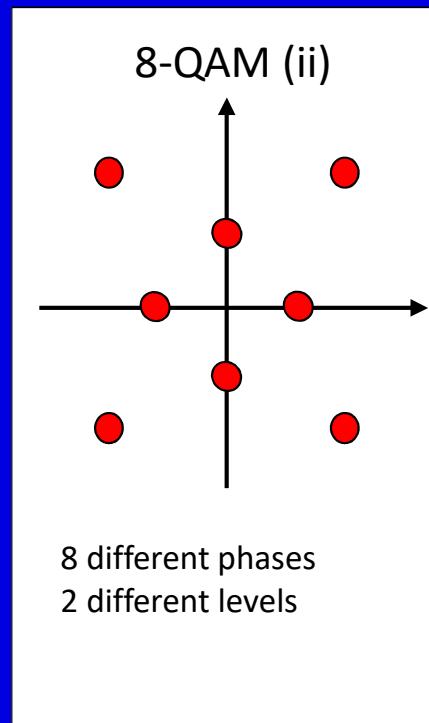
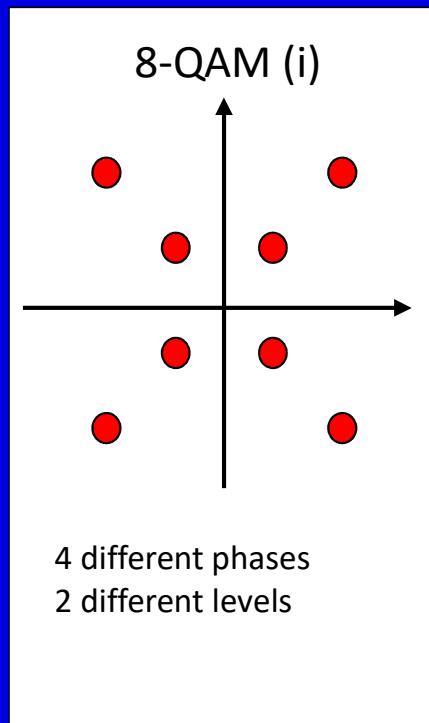
- An ASK symbol with  $M = 2^n$  discrete amplitude levels may carry  $n$  information bits.



- Does ASK has any drawbacks?
  - Amplitude reference is required in the receiver (doesn't apply to 2ASK).
  - Requires linear amplifiers due to the changes in level.
- Are there tricks similar to PSK  $\Rightarrow$  DPSK to avoid the amplitude reference?
  - NO, generally it's not possible to use differential modulation in amplitude shifts.

# QAM - Quadrature Amplitude Modulation

- By combining both phase- and amplitude changes it is possible to accomplish more complex modulating methods.



Are there any drawbacks by using QAM?

- Phase- and amplitude references are required in the receiver.
- Due to change in the level there are no (simple) methods to avoid the references.

Why care about QAM?

- Increased number of bits per modulation symbol.
- Leads to increased data rate and better use of the channel bandwidth.

# Bandwidth Efficiency

- Bandwidth efficiency / information density:

$$D_i = \frac{f_i}{B_i} \quad [\text{bit/sek/Hz}]$$

where  $f_i$  is the information rate (bit/s) and  $B_i$  the corresponding modulation bandwidth (Hz).

- Ideal (minimal) bandwidth at  $M$ -PSK/ $M$ -QAM:

$$B_i = \frac{f_i}{\log_2 M}$$

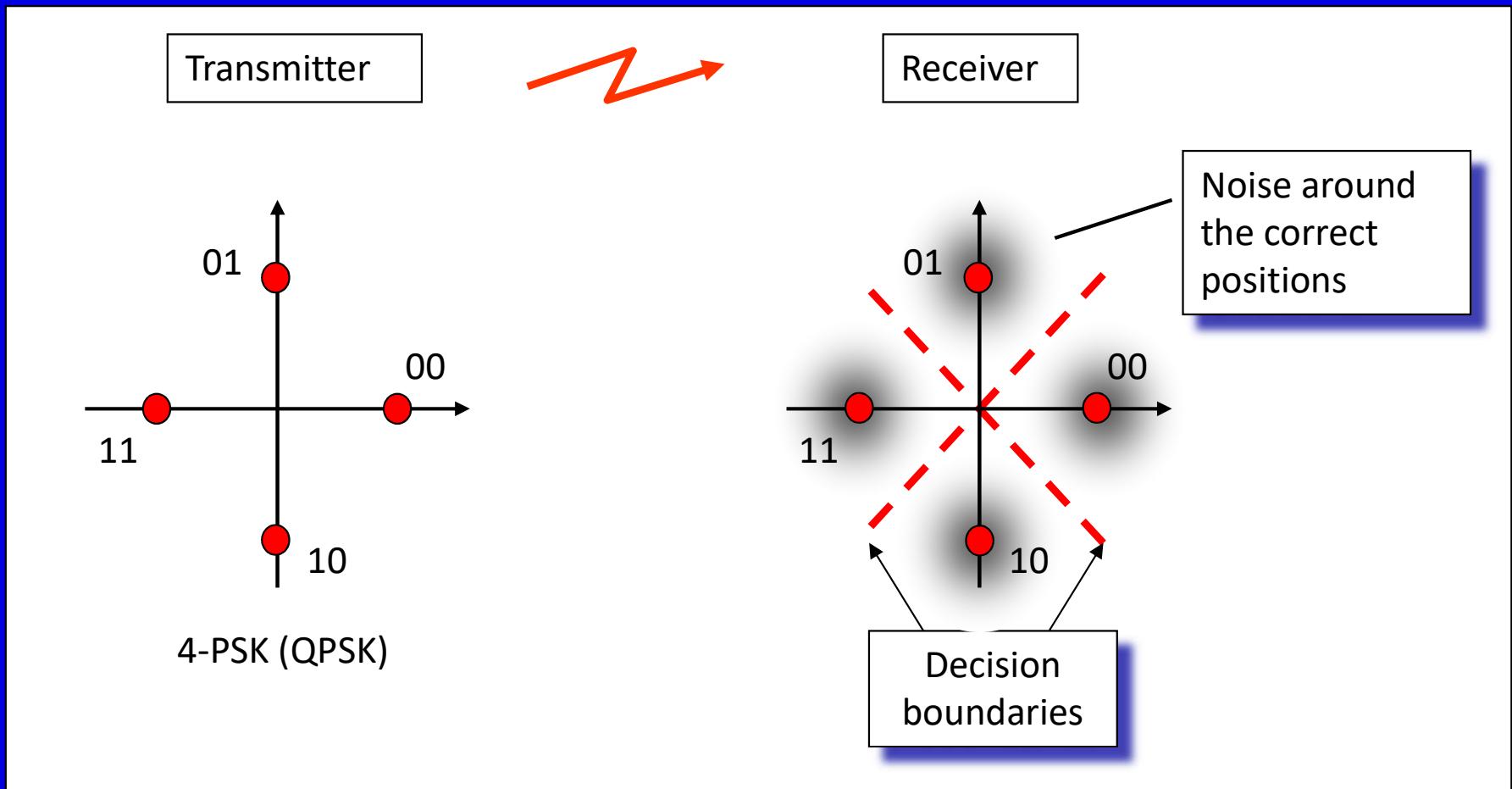
- Combine these equations and we get:

$$D_i = \log_2 M \quad [\text{bit/sek/Hz}]$$

The bandwidth efficiency increases with an increasing  $M$  !

# Noise and Bit Error Rate

- Noise in the receiver system leads to an amount of incorrect detection, a bit error ratio.



# Noise and Bit Error Rate

- The detector characteristics states the relation between the signal quality before the detector and the effective bit error rate (BER).
- The received signal quality is either stated as  $C/N$ ,  $C/N_0$  or  $E_b/N_0$ , where  $C$  is the received power [Watt],  $E_b$  is the received energy per bit [Joule/bit] and  $N_0$  is the noise density [Watt/Hz].
- By using the notation from the bandwidth efficiency we get:

compare to  
the modulation  
gain!

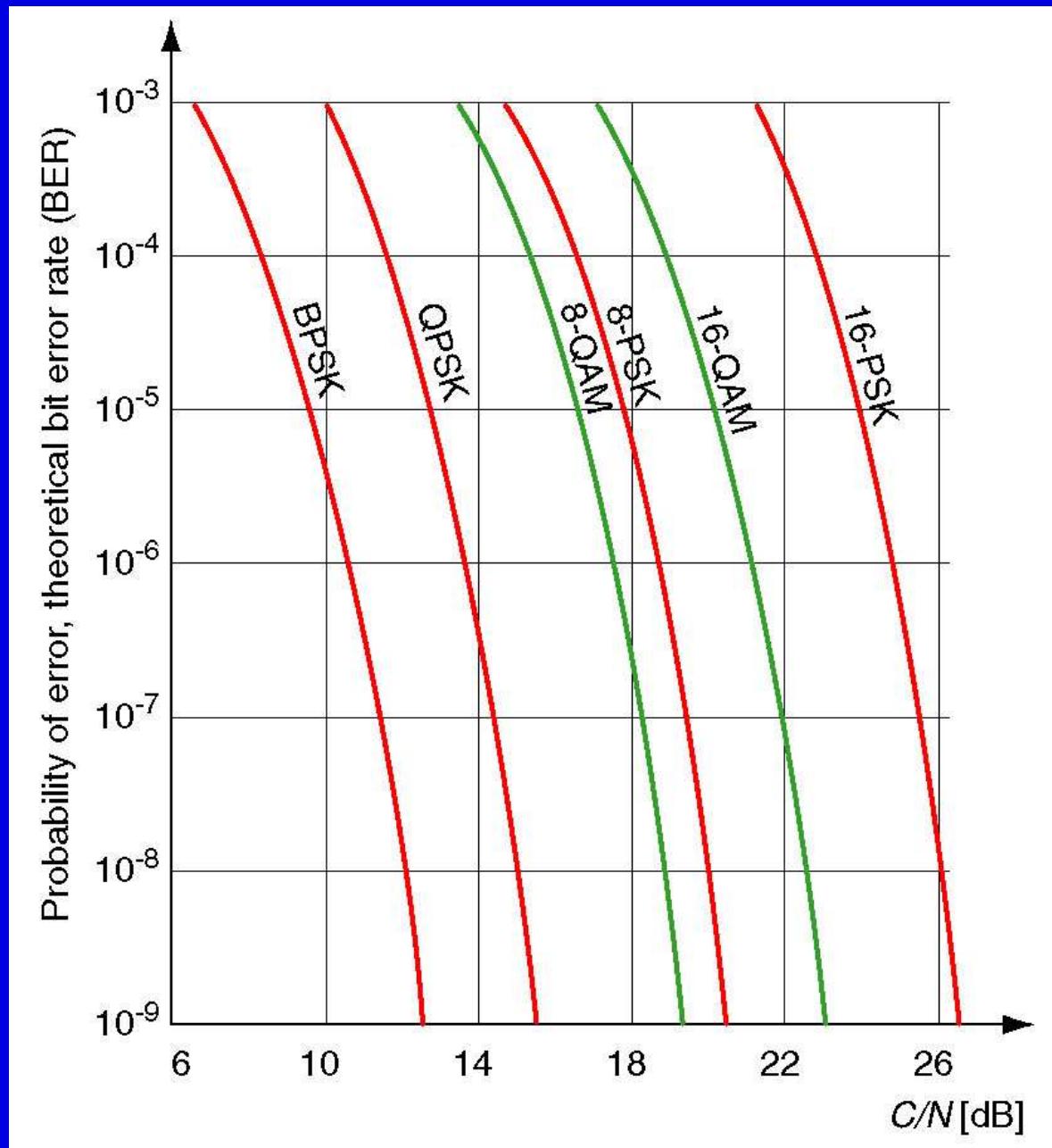
$$C = E_b f_i$$
$$N = N_0 B_i$$



$$\frac{C}{N} = \frac{E_b}{N_0} \frac{f_i}{B_i}$$
$$\frac{C}{N_0} = \frac{E_b}{N_0} f_i$$

These equations state the relation between  $E_b/N_0$  that is common in BER charts and the corresponding received power  $C$ .

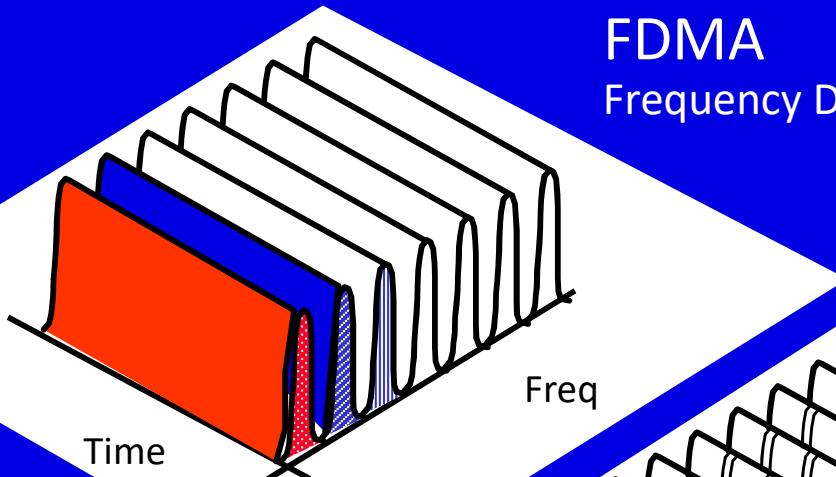
# Noise and Bit Error Rate



# Multiple-Access Techniques

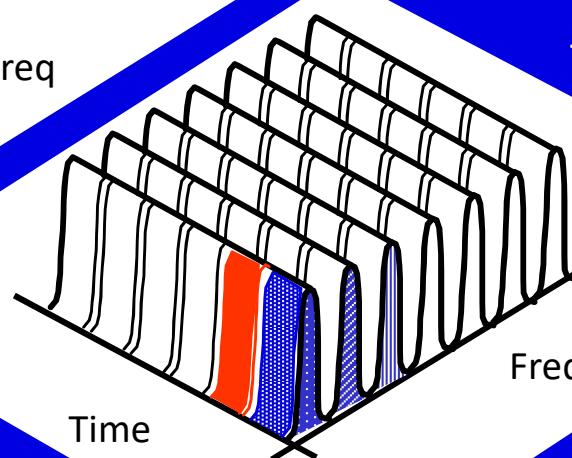
FDMA

Frequency Division Multiple Access



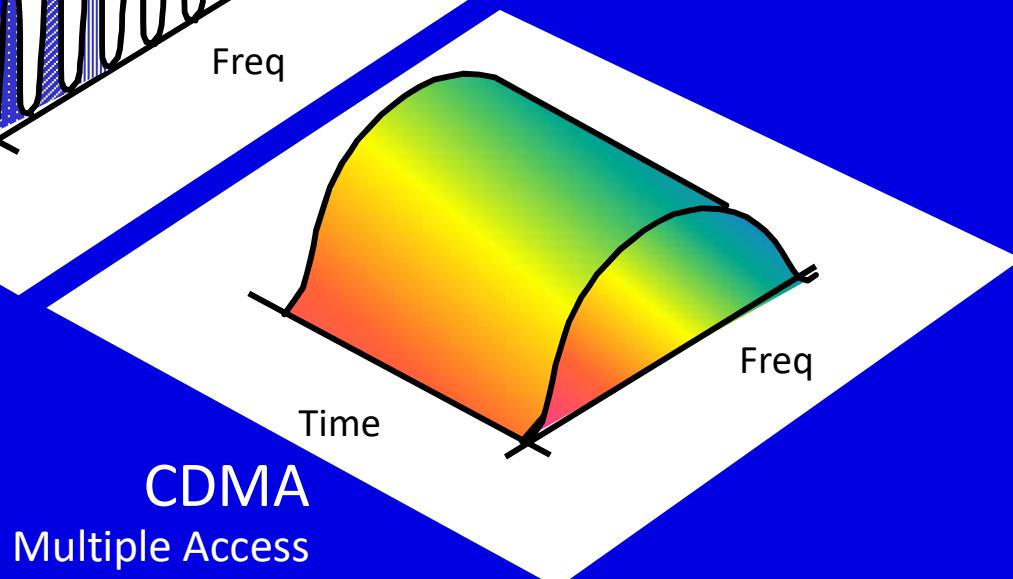
TDMA

Time Division Multiple Access



CDMA

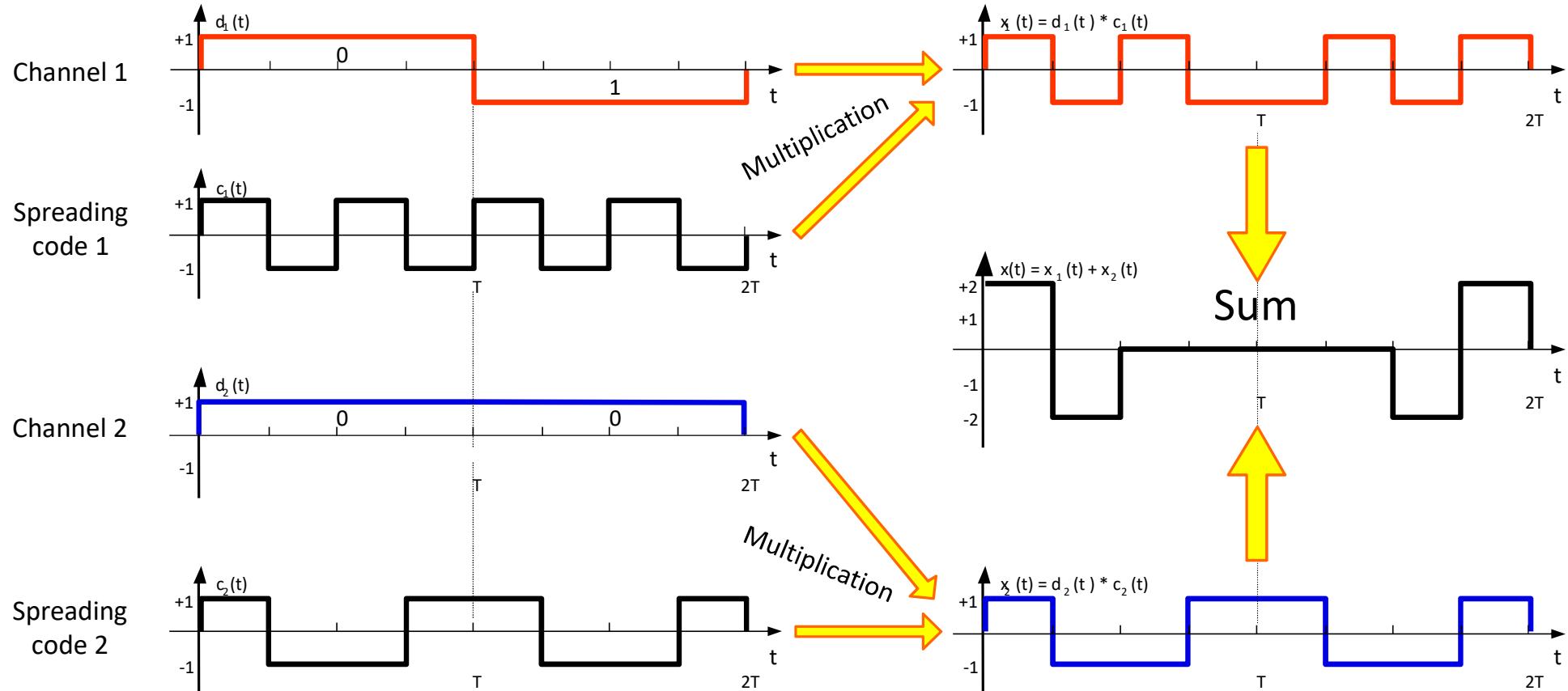
Code Division Multiple Access



# Spreading of Two Sources

Example of a transmitter

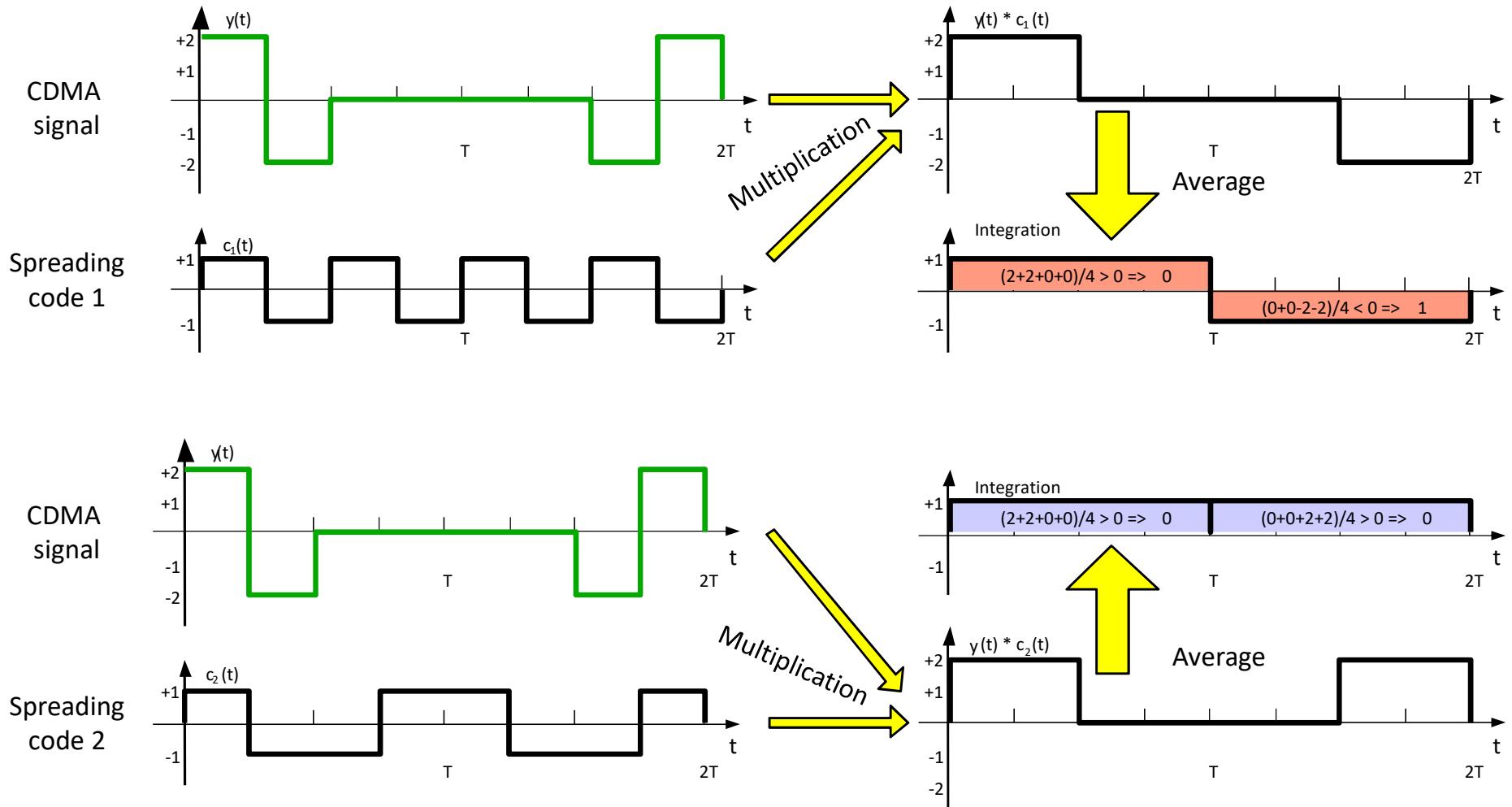
$0 \rightarrow +1$   
 $1 \rightarrow -1$



# De-Spreading of Two Sources

## Example of a receiver

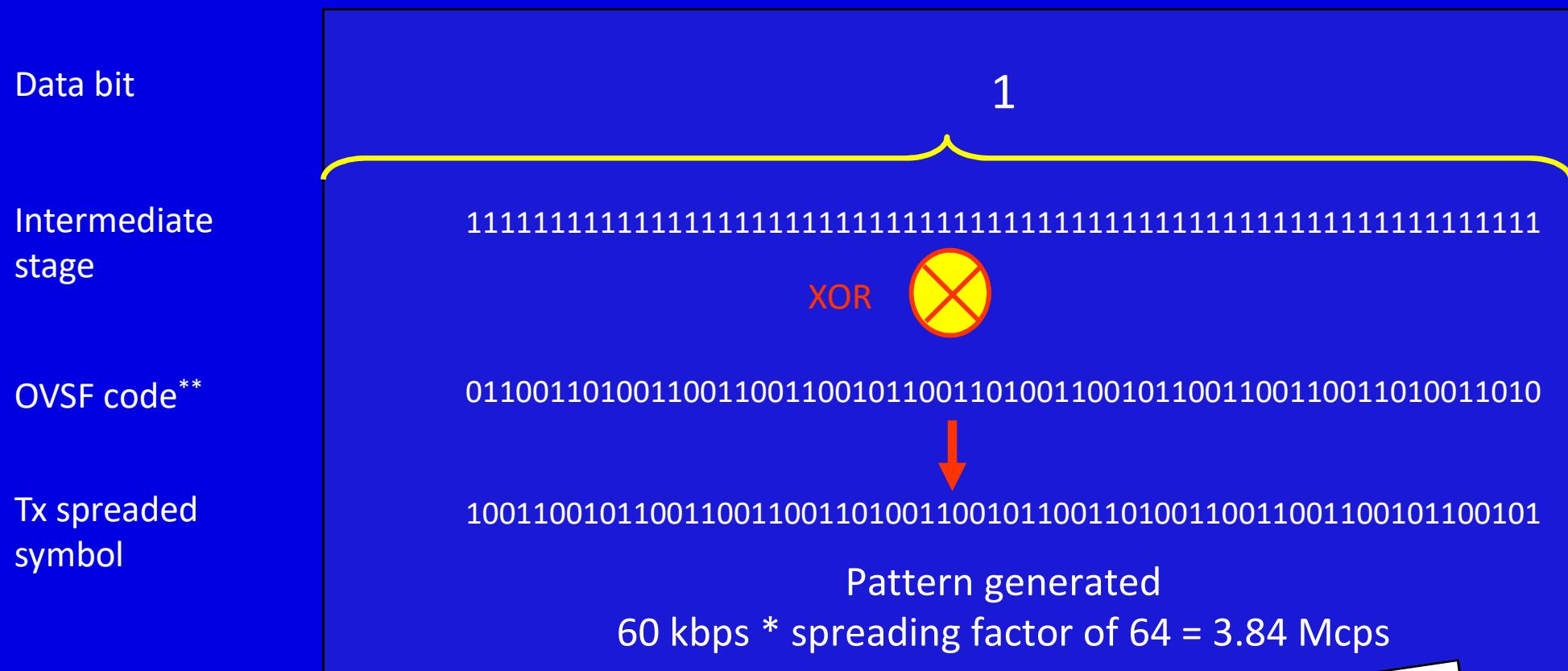
$0 \rightarrow +1$   
 $1 \rightarrow -1$



# Spreading Code

3GPP\* chip rate: 3.84 Mcps

## Example: 60 kbps data rate

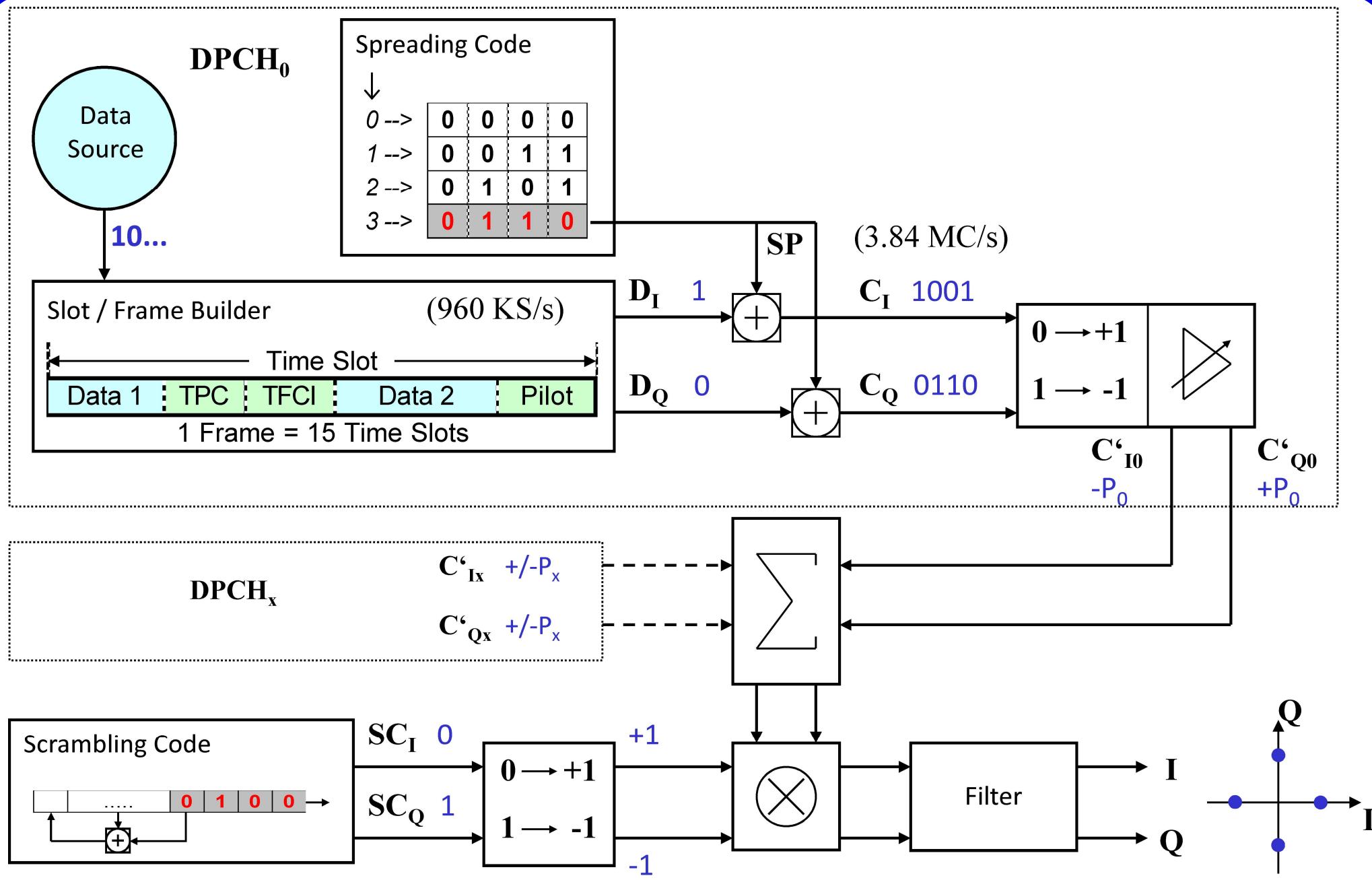


\* 3rd Generation Partnership Project

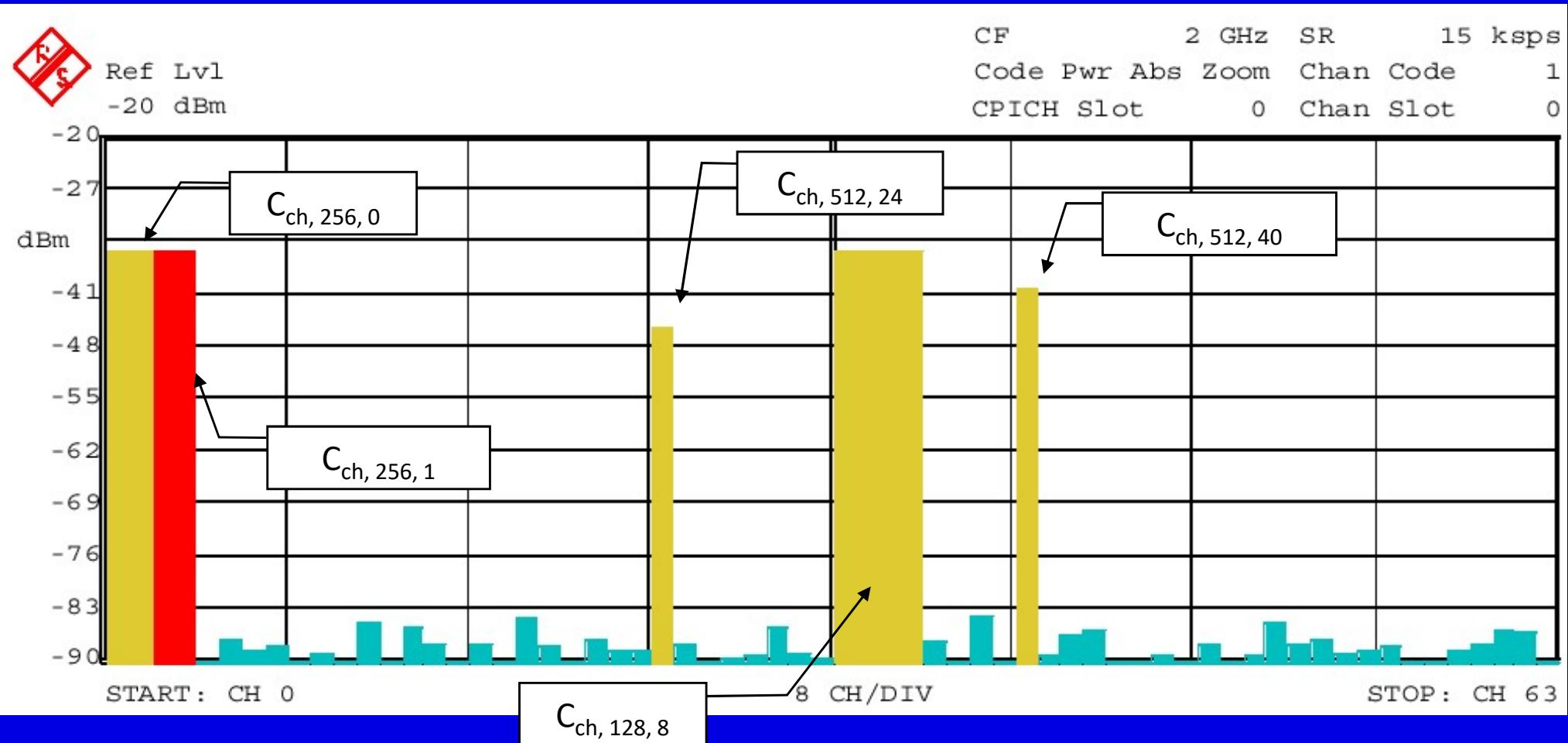
## \*\* Orthogonal Variable Spreading Factor

It needs 64 bits (1 symbol) to transmit 1 original bit!

# Channel Coding

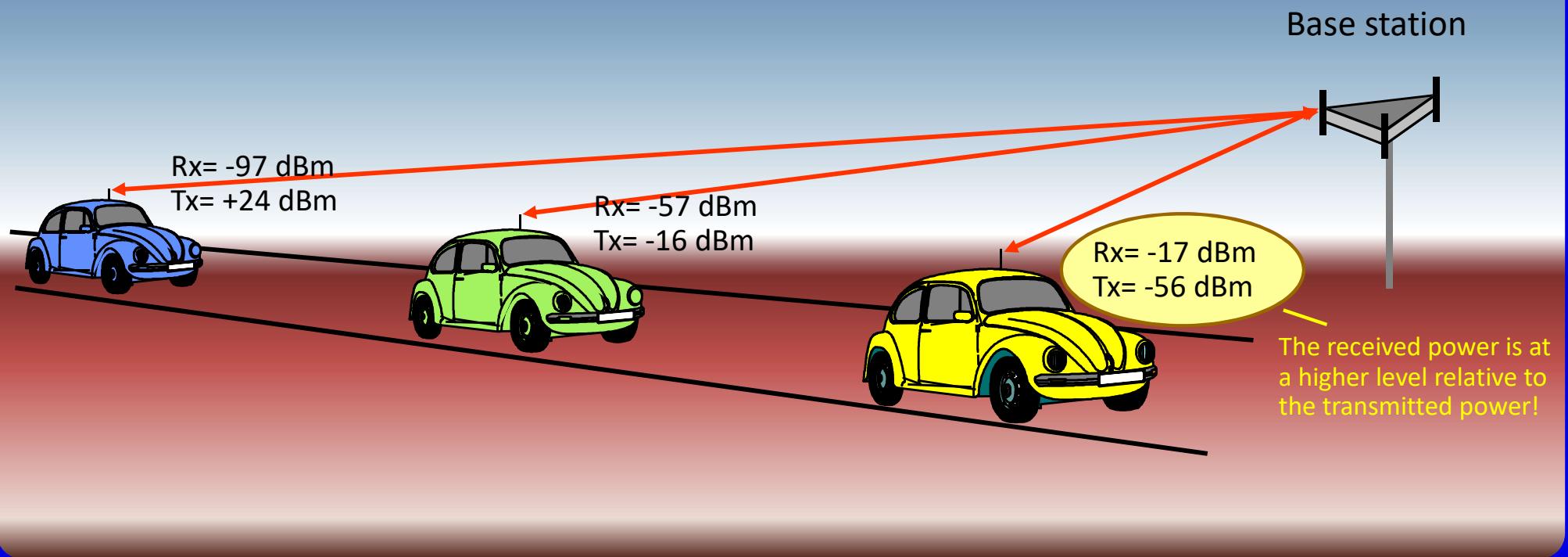


# Code Domain Measurements



- At CDMA it is important that all channels in the code domain are received at equal levels
- Therefore is a dynamic power control required in the transmitter

# “Open Loop” Power Control



- Fast control: 10 dB setting at 10 - 100 ms
- A “closed loop” power control is also used that is more accurate but causes some delay

# Multi-carrier OFDM... Why?

- What happens if you speak rapidly in a large open stone building?

ECHOES ... ECHOES ... ECHOES !

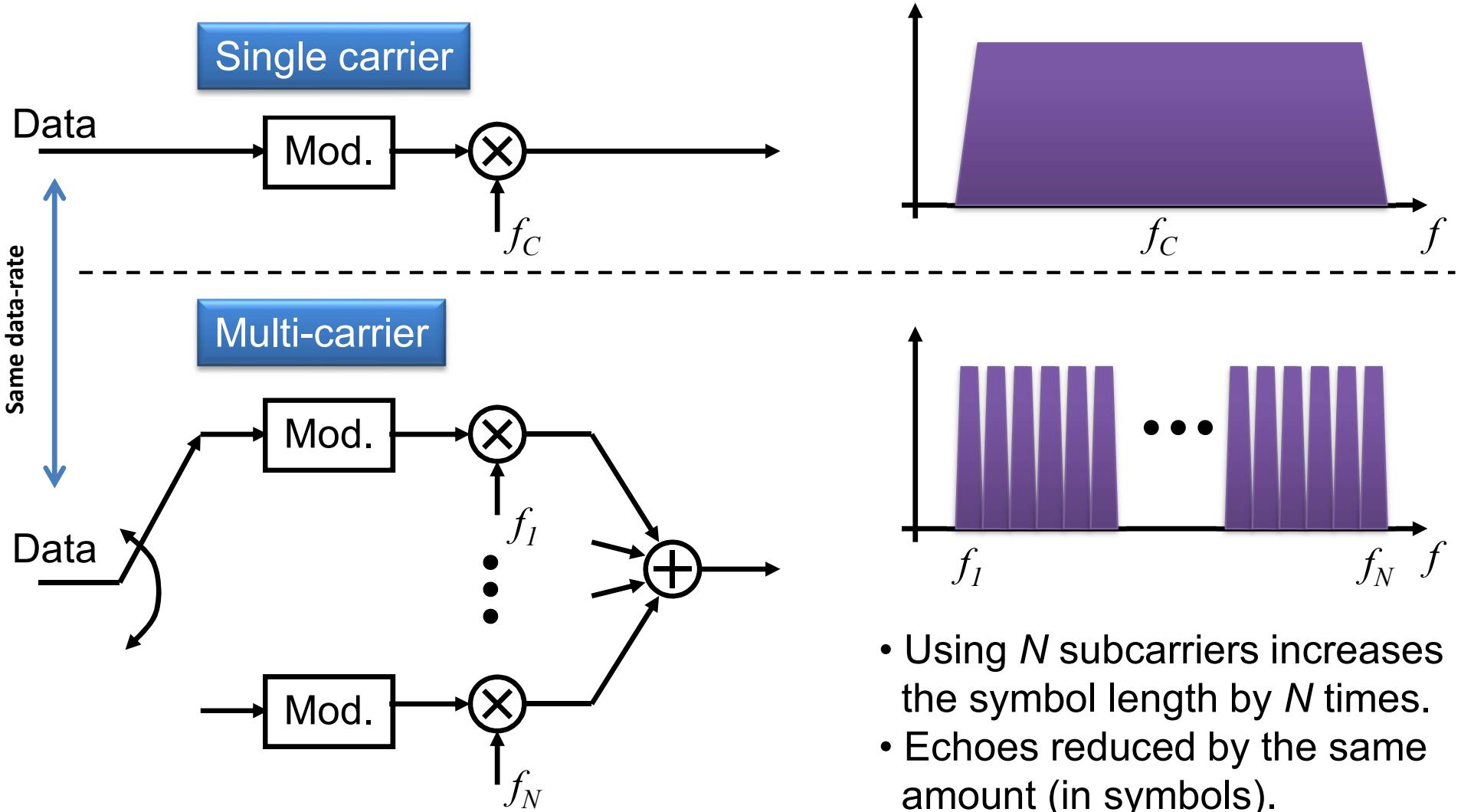
How do you solve the problem?

SPEAK SLOWER!

THIS IS THE ESSENCE  
OF MULTI-CARRIER!  
Slower transmission on each  
carrier, but use many carriers



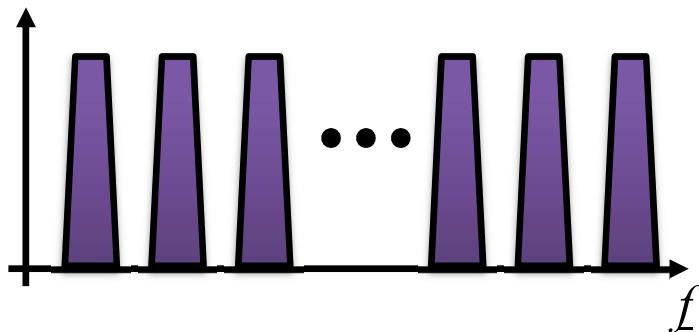
# Single/multi-carrier



- Using  $N$  subcarriers increases the symbol length by  $N$  times.
- Echoes reduced by the same amount (in symbols).

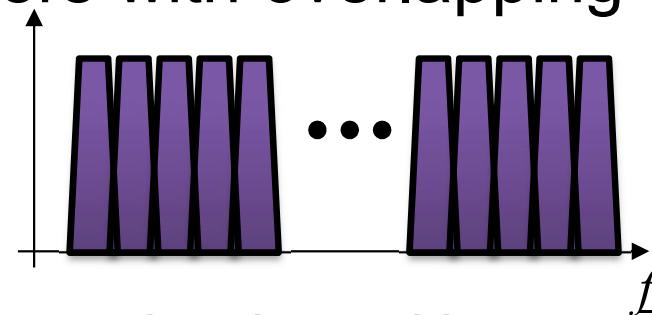
# History and evolution [1]

**1950's:** Few subcarriers, with non-overlapping spectra



- Military systems, e.g. the Kineplex-modem

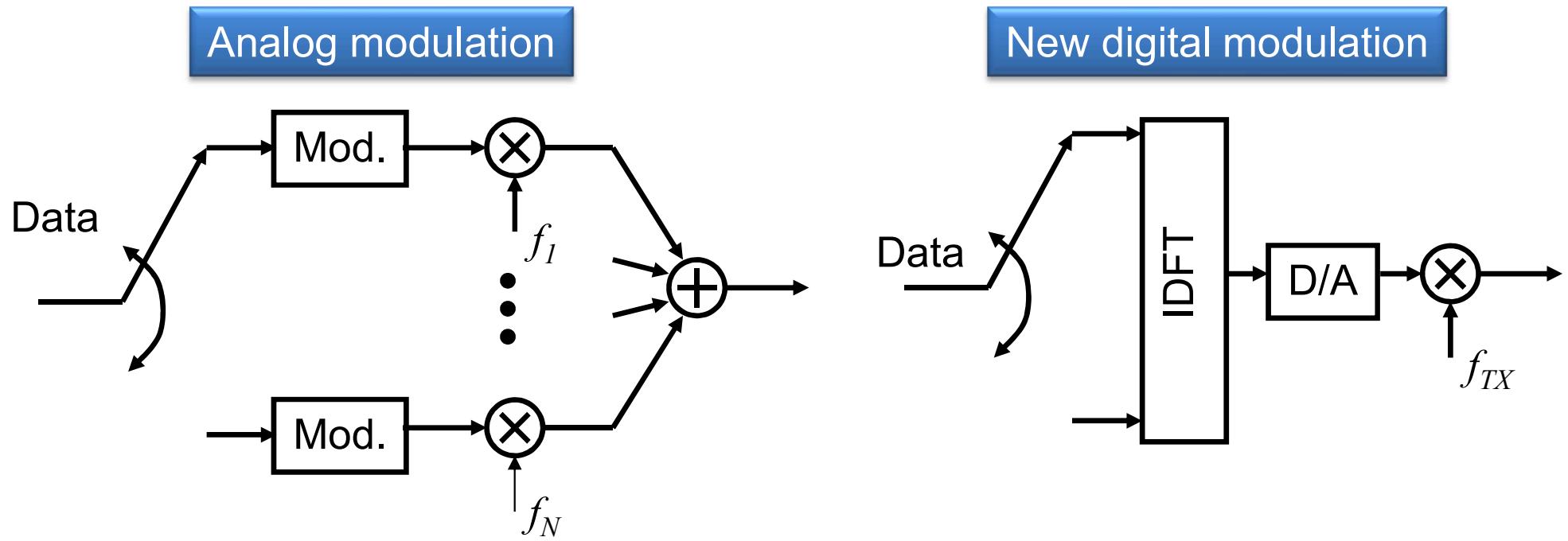
**1960's:** Subcarriers with overlapping spectra



Increased subchannel density and increased data rate.

# History and evolution [2]

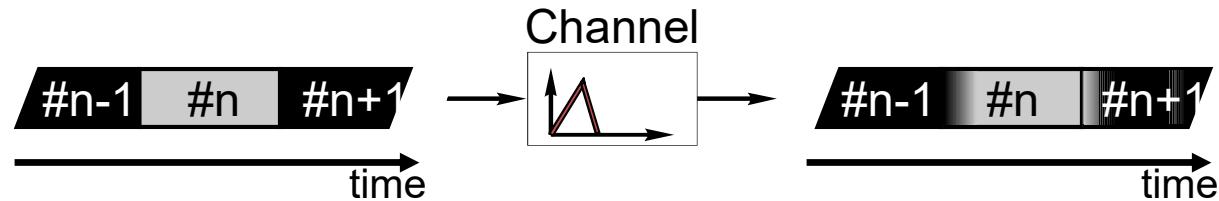
1970's: Digital modulation of subcarriers



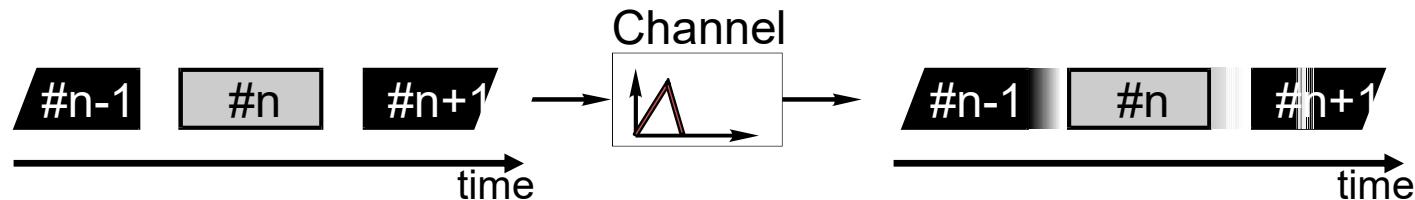
# History and evolution [3]

1980's: Improved digital circuits increases interest

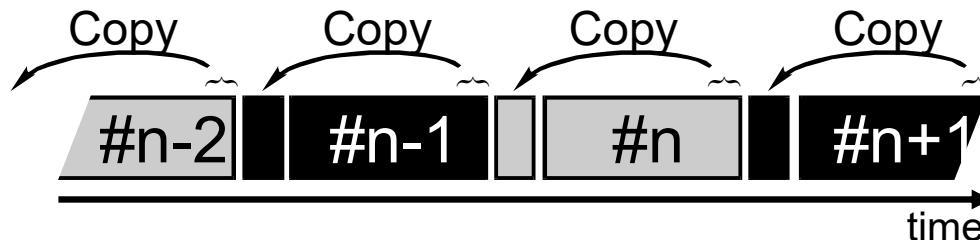
No guard interval => Interference between both subcarriers and symbols



Guard interval => No interference between symbols



Cyclic prefix => No interference between neither subcarriers nor symbols

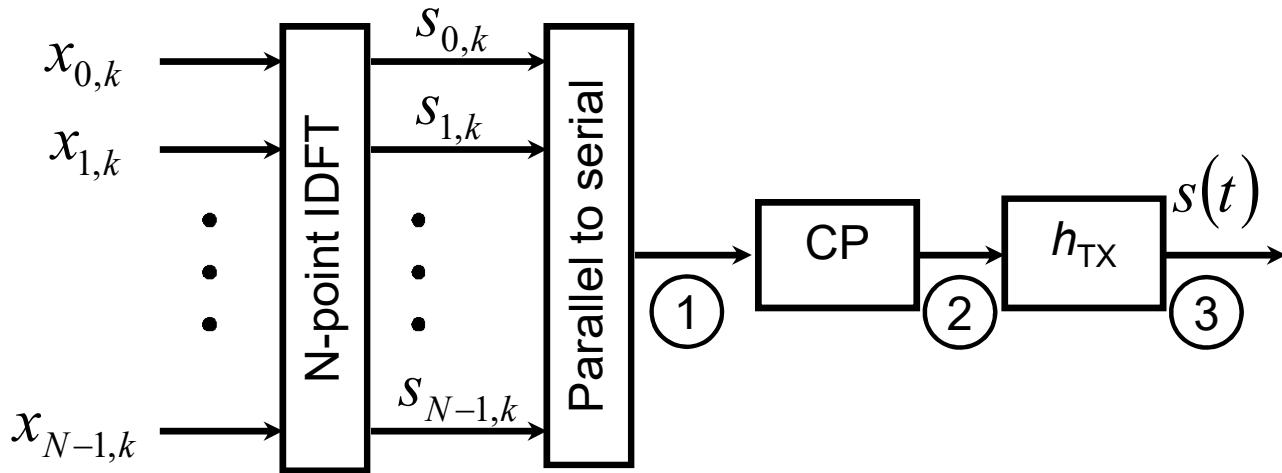


# History and evolution [4]

- **1990's:** Commercial applications appear
  - Increased interest for OFDM in wireless applications
  - First applications in broadcasting (Audio/Video)
  - One of the candidates for UMTS (Beta proposal)
  - Applied in wireless LANs
- **2000's:** One of the really hot technologies
  - > 50 Mbps WLANs (based on OFDM) hit the mass market
  - OFDM is one of the candidate technologies when improving and moving beyond current 3G systems – LTE!

# Transmitters and receivers

## An N-subcarrier transmitter

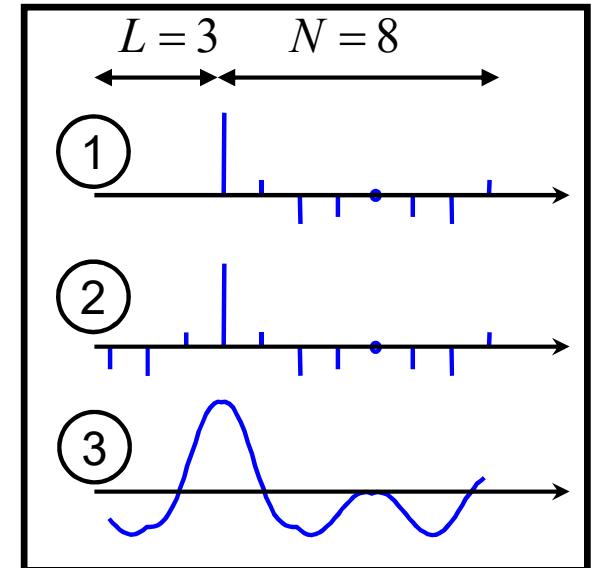


$k$	– symbol
$m$	– sample
$n$	– subcarrier
$L$	– CP length
$T_{\text{samp}}$	– sampling period
$h_{\text{TX}}$	– TX filter

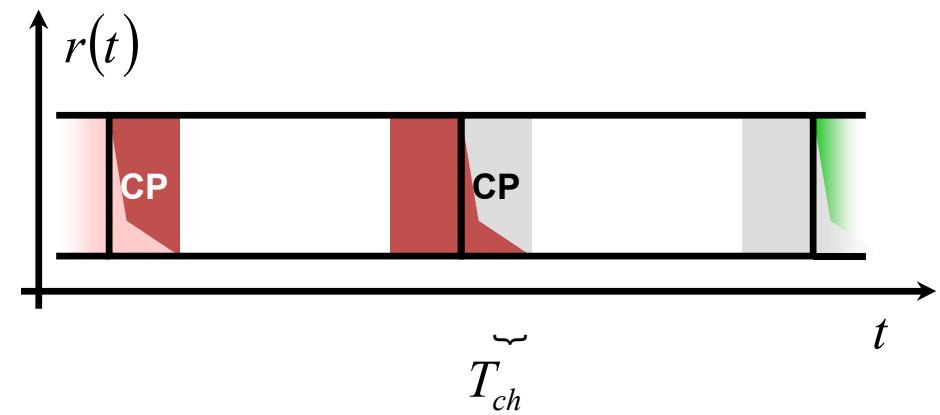
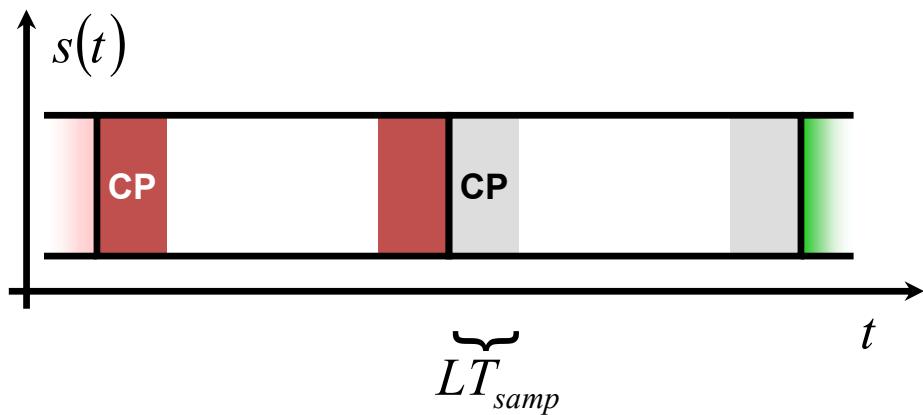
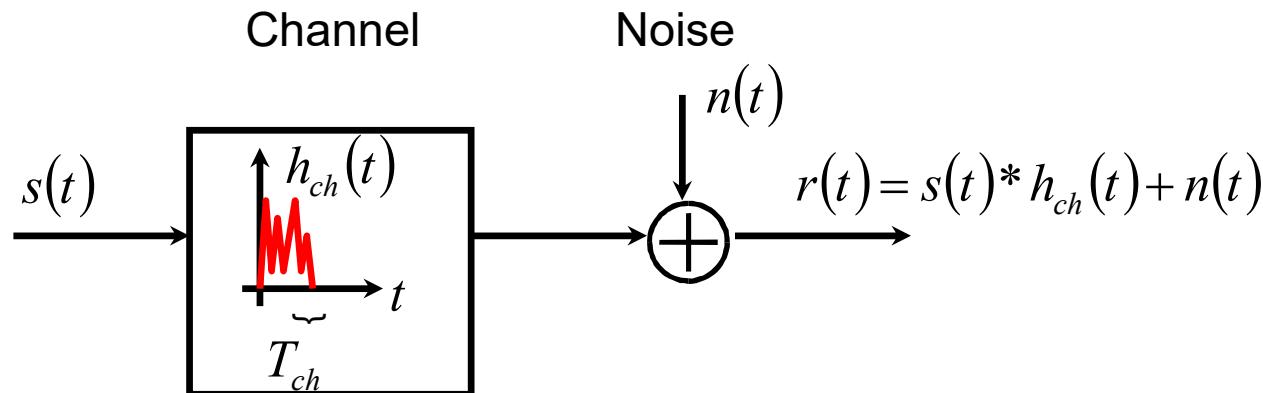
$$\text{N-point IDFT: } s_{m,k} = \frac{1}{\sqrt{N}} \sum_{n=0}^{N-1} x_{n,k} \exp\left(j2\pi \frac{mn}{N}\right) \text{ for } 0 \leq m \leq N-1$$

Adding CP:  $s_{m,k} = s_{N+m,k}$  for  $-L \leq m \leq -1$

$$\text{TX filtering: } s(t) = h_{\text{TX}}(t) * \left( \sum_k \sum_{m=-L}^{N-1} s_{m,k} \delta(t - k(N+L)T_{\text{samp}} - mT_{\text{samp}}) \right)$$



# Transmitters and receivers ... through the channel ...

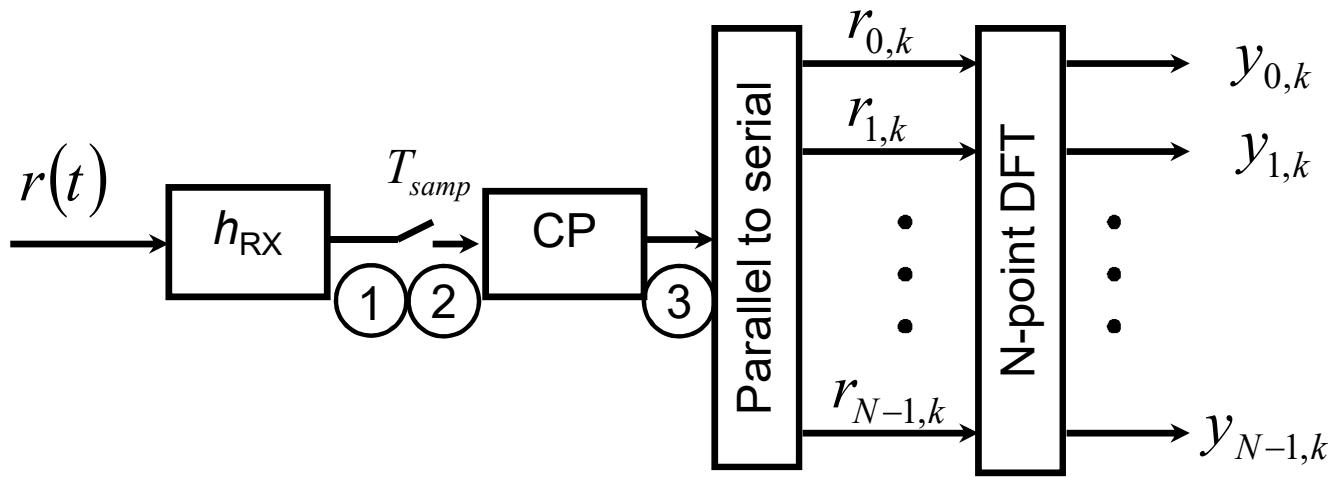


As long as the CP is longer than the delay spread of the channel,  $L T_{samp} > T_{ch}$ , it will absorb the ISI.

By removing the CP in the receiver, the transmission becomes ISI free.

# Transmitters and receivers

## N-subcarrier receiver



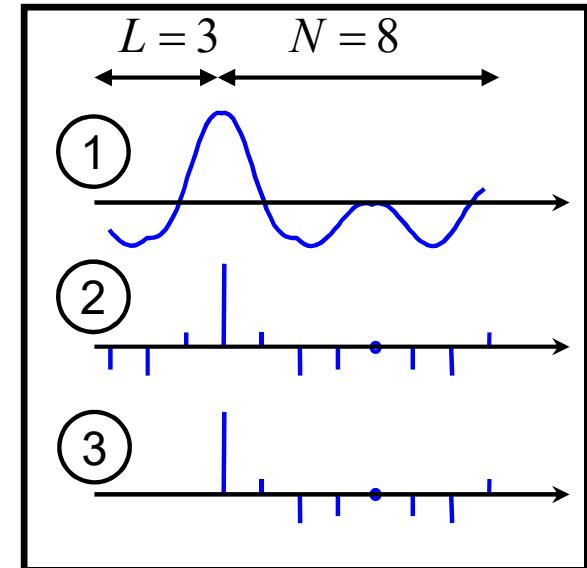
$q$	– symbol
$p$	– sample
$n$	– subcarrier
$L$	– CP length
$T_{\text{samp}}$	– sampling period
$h_{RX}$	– RX filter

RX filtering:  $z'(t) = h_{RX}(t) * r(t)$

Sampling:  $z'_k = z'(kT_{\text{samp}})$

Removing CP:  $r_{p,q} = z'_{q(N+L)+p}$  for  $0 \leq p \leq N-1$

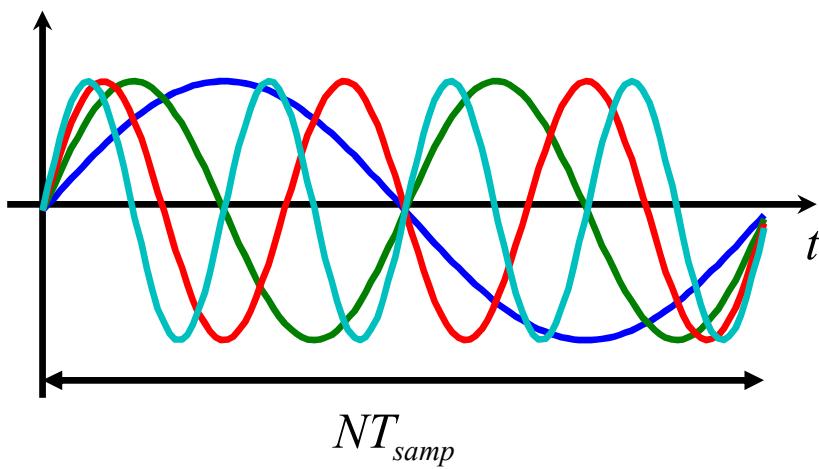
N-point DFT:  $y_{n,q} = \frac{1}{\sqrt{N}} \sum_{p=0}^{N-1} r_{p,q} \exp\left(-j2\pi \frac{np}{N}\right)$  for  $0 \leq n \leq N-1$



# Transmitters and receivers

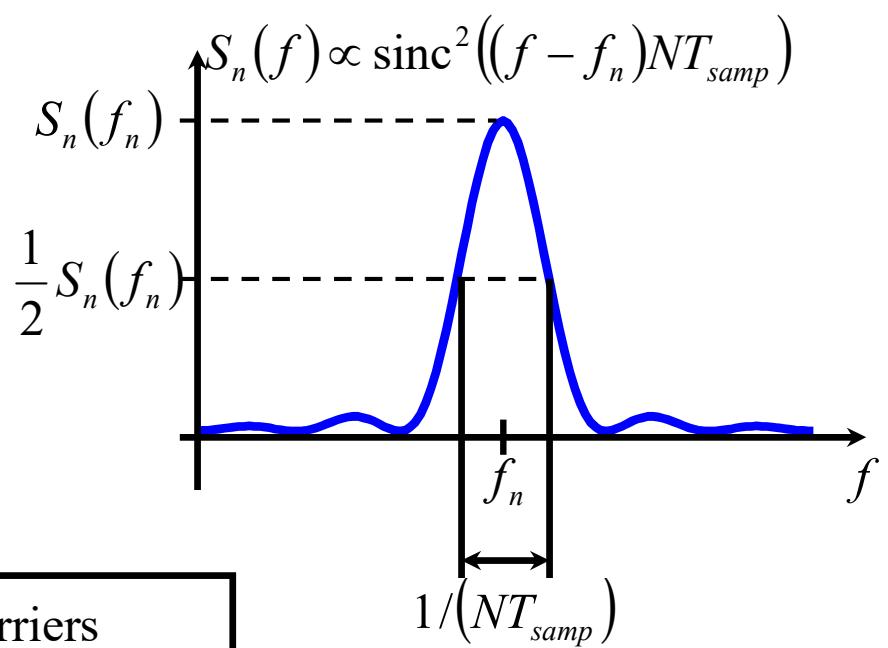
## Modulation spectrum [1]

Transmitted OFDM symbol decomposed  
into different subcarriers (ideal case,  
4 subcarriers shown, no CP)



$N$	-	Subcarriers
$T_{\text{samp}}$	-	Sampling period
$\text{sinc}(x) = \frac{\sin \pi x}{\pi x}$		

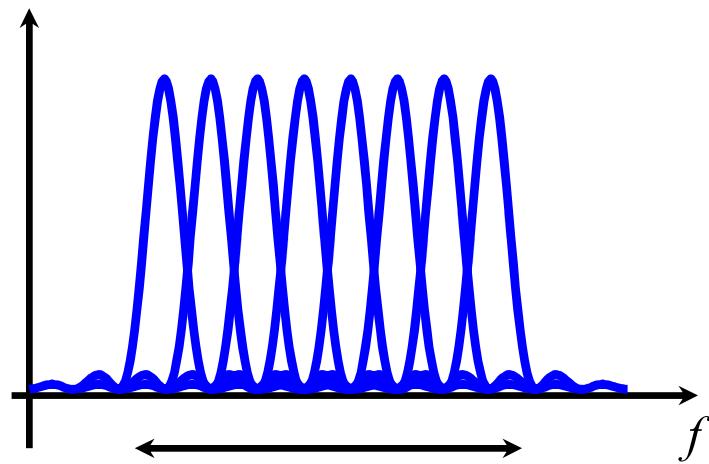
Power spectrum of one subcarrier  
transmitted at  $f_n$  Hz.



# Transmitters and receivers

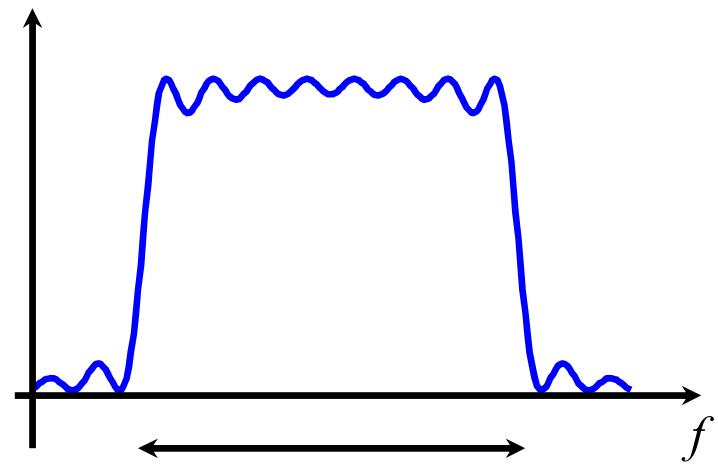
## Modulation spectrum [2]

The distance between each subcarrier becomes  $1/(NT_{\text{samp}})$  which is the same as the 3 dB bandwidth of the individual subcarriers. Using all  $N$  subcarriers (8 in this case) we get:



$$B = N \times 1/(NT_{\text{samp}}) = 1/T_{\text{samp}}$$

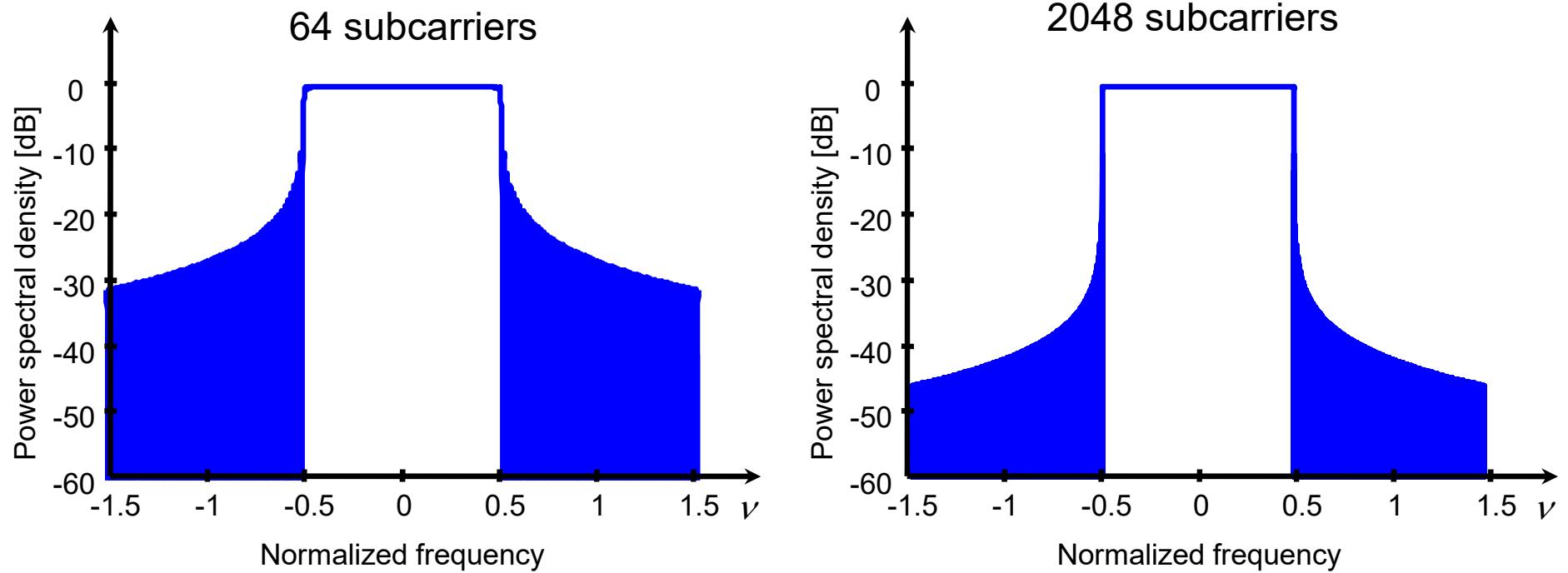
The total modulation spectrum is a sum of the individual subcarrier spectra (assuming independent data on them).



$$B = 1/T_{\text{samp}}$$

# Transmitters and receivers

## Modulation spectrum [3]

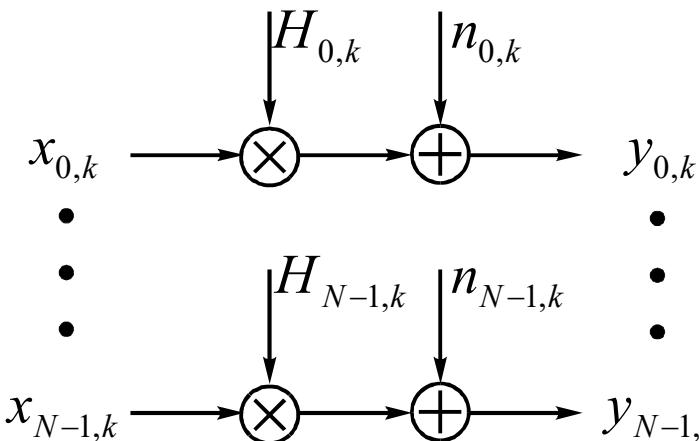


Normalized freq.:  
 $\nu = T_{\text{samp}} f = f / B$

# Transmitters and receivers

## Simplified model

Simplified model under ideal conditions  
(no fading and sufficient CP)



Total filter in the signal path:

$$h_{signal}(t) = h_{TX}(t) * h(t) * h_{RX}(t)$$

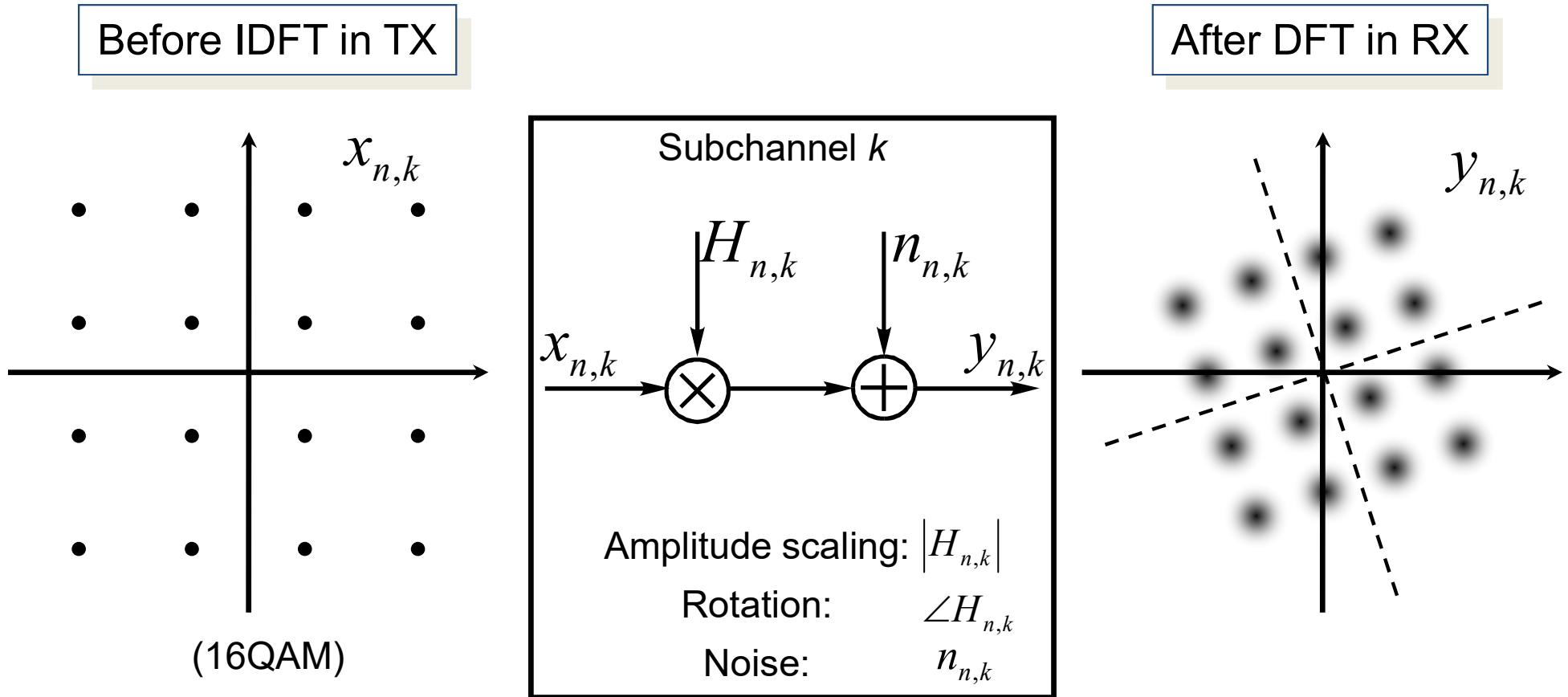
$$H_{signal}(f) = H_{TX}(f) * H(f) * H_{RX}(f)$$

Given that subcarrier  $n$  is transmitted at frequency  $f_n$   
the attenuations become:

$$H_{n,k} = H_{signal}(f_n)$$

# Transmitters and receivers

## Focus on one subchannel



- Simple equalization of each subchannel: Back-rotate and scale