

# Wireless Sensor Networks

## 1. Basics

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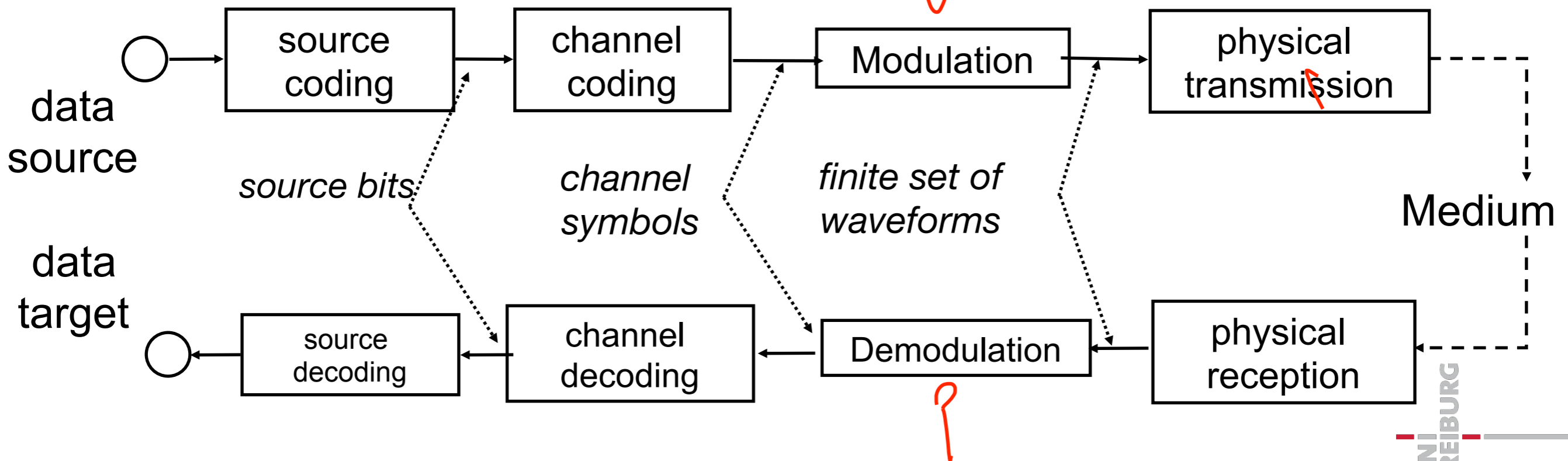
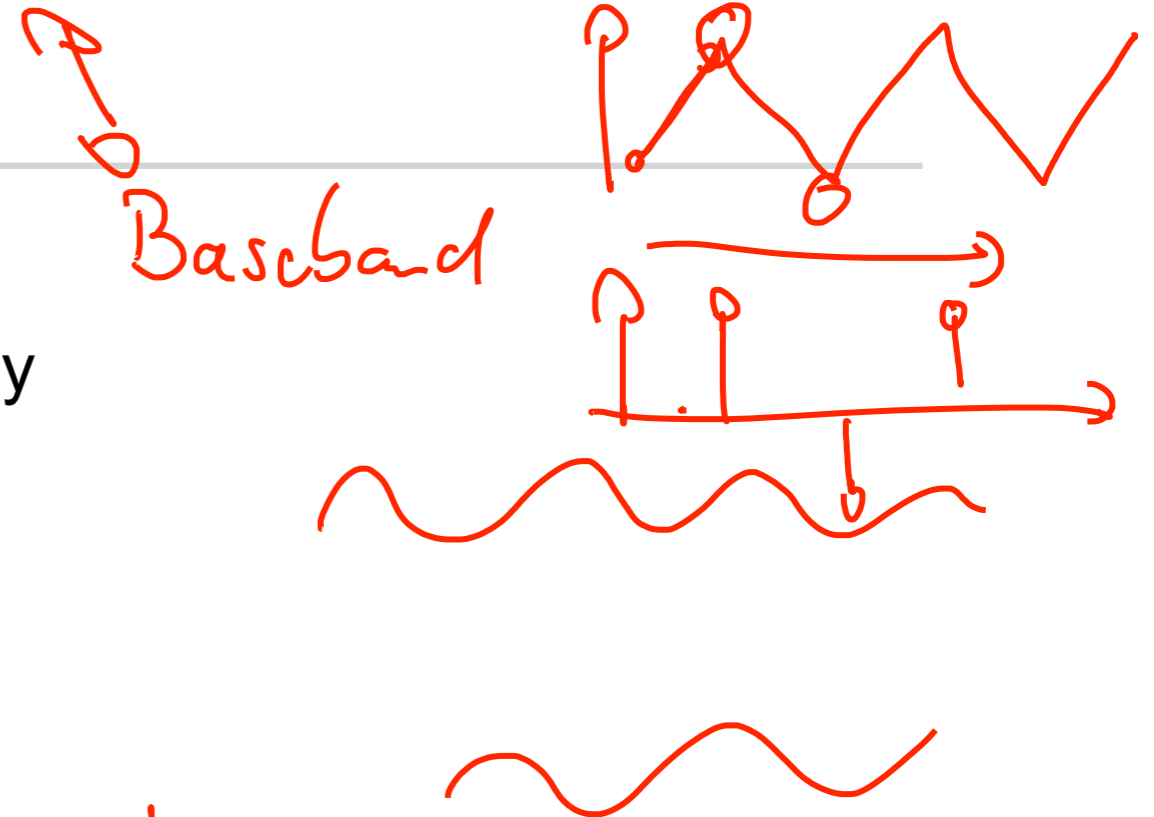
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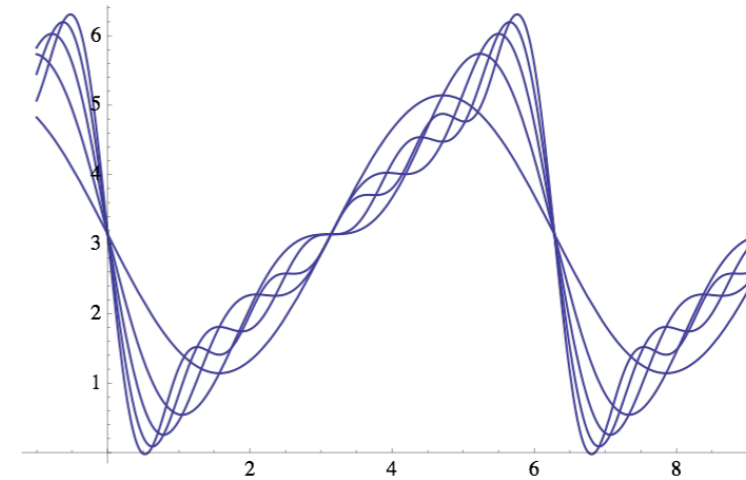
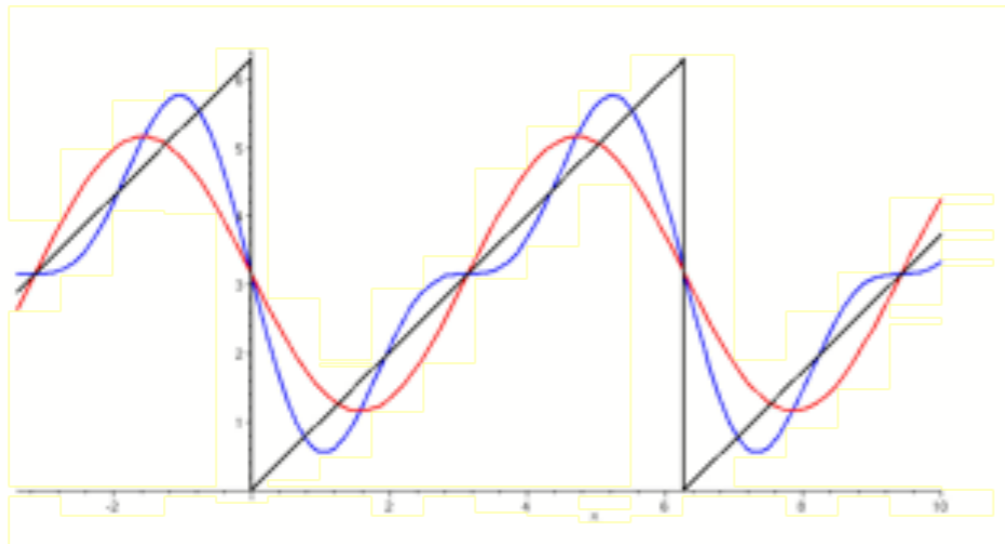
# Structure of a *Broadband* Digital transmission

## ■ MODOulation/DEMODulation

- Translation of the channel symbols by
  - amplitude modulation
  - phase modulation
  - frequency modulation
  - or a combination thereof



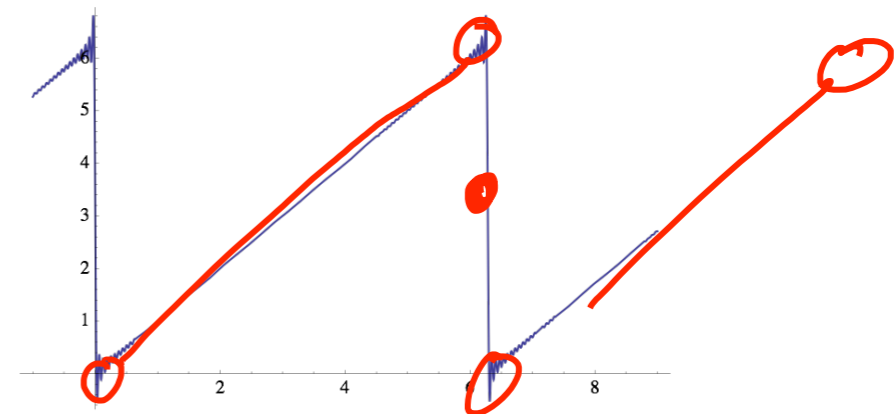
# Computation of Fourier Coefficients



$$f(x) = x, \text{ für } 0 < x < 2\pi$$

$$f(x) = \pi - 2 \left( \frac{\sin x}{1} + \frac{\sin 2x}{2} + \frac{\sin 3x}{3} + \dots \right)$$

$\frac{d \sin x}{dx} = \cos x$   
 $\frac{d \cos x}{dx} = -\sin x \rightarrow -\cos x \rightarrow \sin x$



- Theorem of Fourier for period  $T=1/f$ :
  - The coefficients  $c$ ,  $a_n$ ,  $b_n$  are then obtained as follows

$$g(t) = \frac{a_0}{2} + \sum_{k=1}^{\infty} a_k \cos(2\pi k f t) + b_k \sin(2\pi k f t)$$

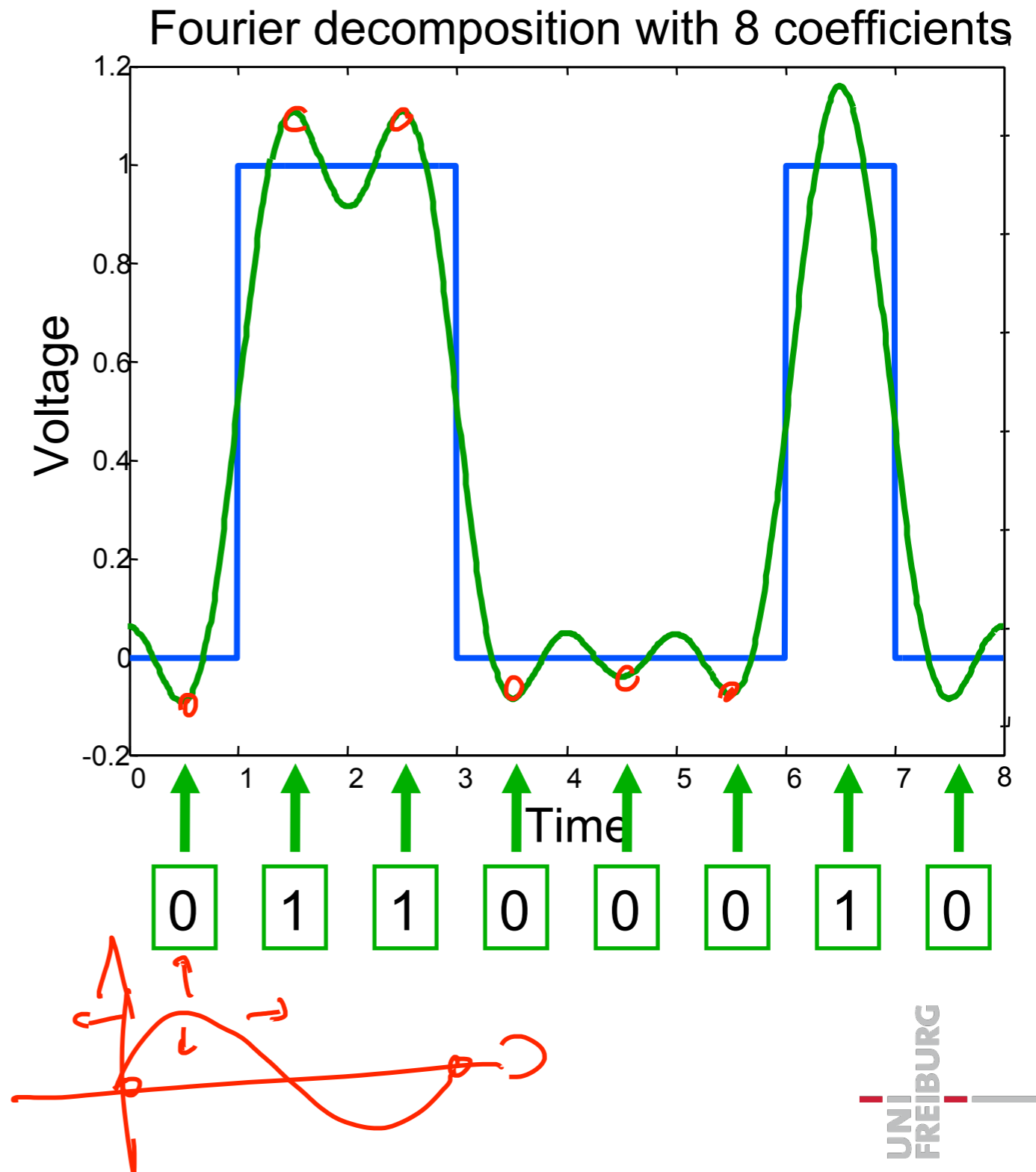
$$a_k = \frac{2}{T} \int_0^T g(t) \cos(2\pi n f t) dt$$

$$b_k = \frac{2}{T} \int_0^T g(t) \sin(2\pi n f t) dt$$

- The sum of squares of the  $k$ -th terms is proportional to the energy consumed in this frequency:
 
$$(a_k)^2 + (b_k)^2$$

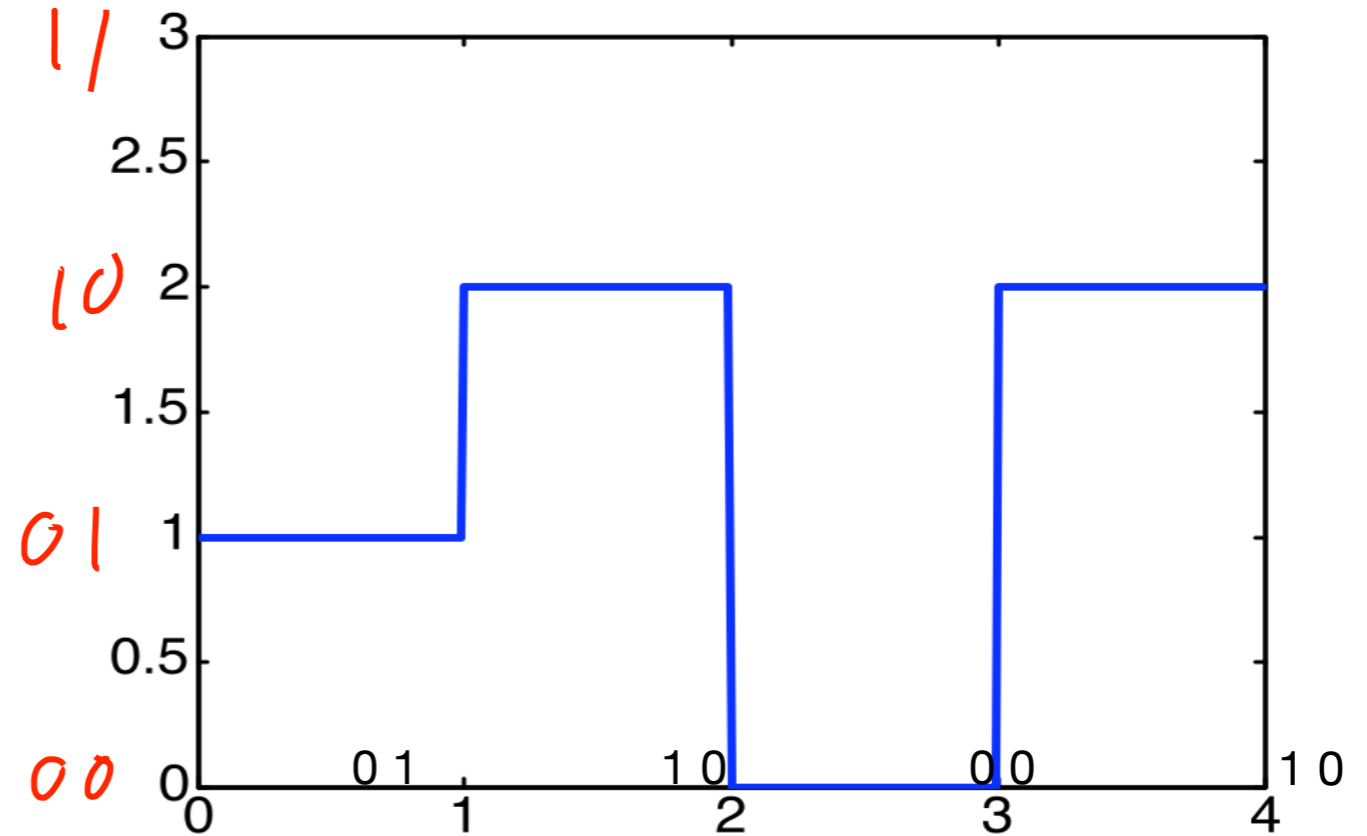
# How often do you measure?

- How many measurements are necessary
  - to determine a Fourier transform to the  $k$ -th component, exactly?
- Nyquist-Shannon sampling theorem
  - To reconstruct a continuous band-limited signal with a maximum frequency  $f_{\max}$  you need at least a sampling frequency of  $2 f_{\max}$ .



# Symbols and Bits

- For data transmission instead of bits can also be used symbols
  - E.g. 4 Symbols: A, B, C, D with
    - A = 00, B = 01, C = 10, D = 11
- Symbols
  - Measured in baud
  - Number of symbols per second
- Data rate
  - Measured in bits per second (bit / s)
  - Number of bits per second
- Example
  - 2400 bit/s modem is 600 baud (uses 16 symbols)



- Idea

- Focusing on the ideal frequency of the medium
- Using a sine wave as the carrier wave signals

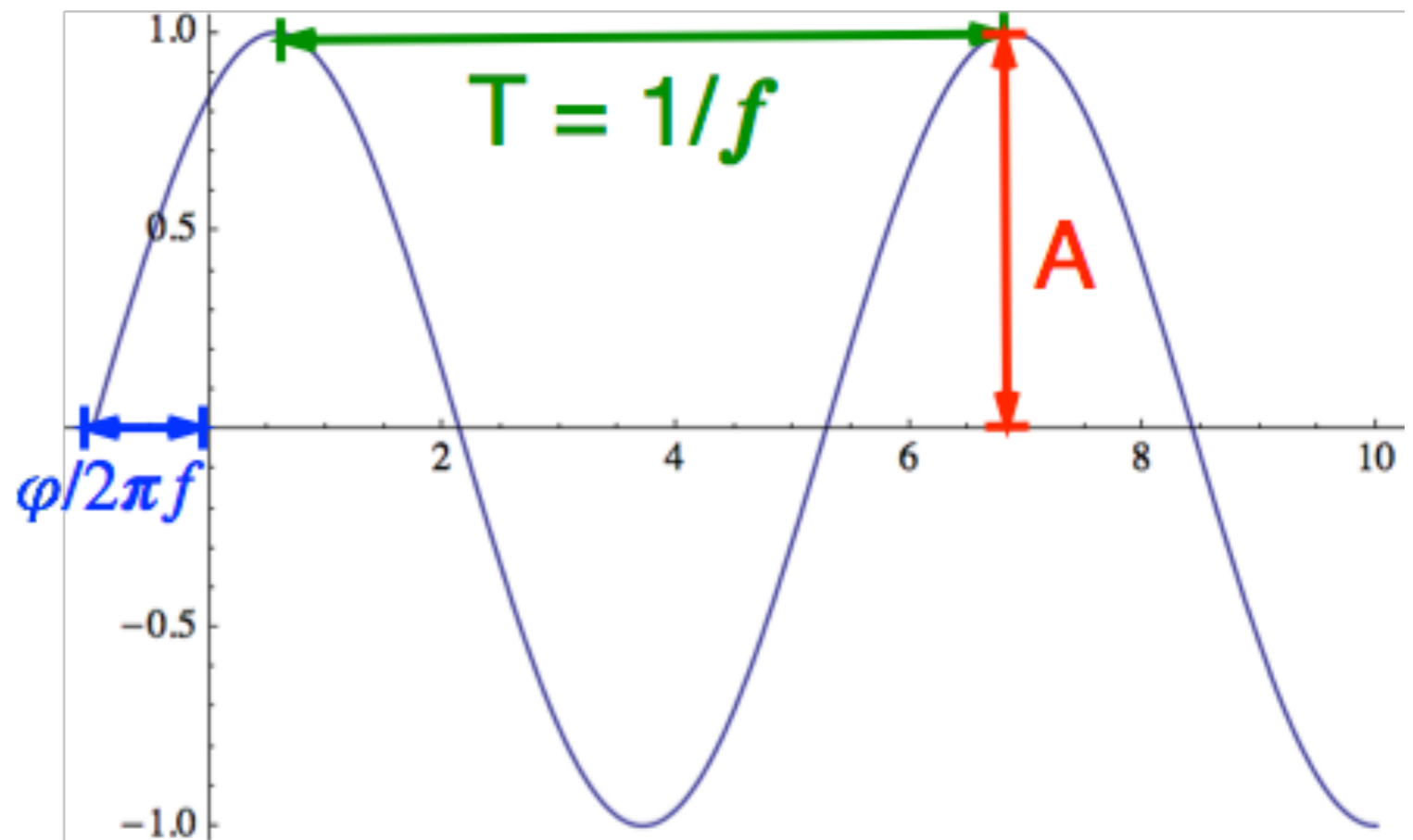
- A sine wave has no information

- the sine curve continuously (modulated) changes for data transmission,
- implies spectral widening (more frequencies in the Fourier analysis)

- The following parameters can be changed:

- Amplitude  $A$
- Frequency  $f=1/T$
- Phase  $\varphi$

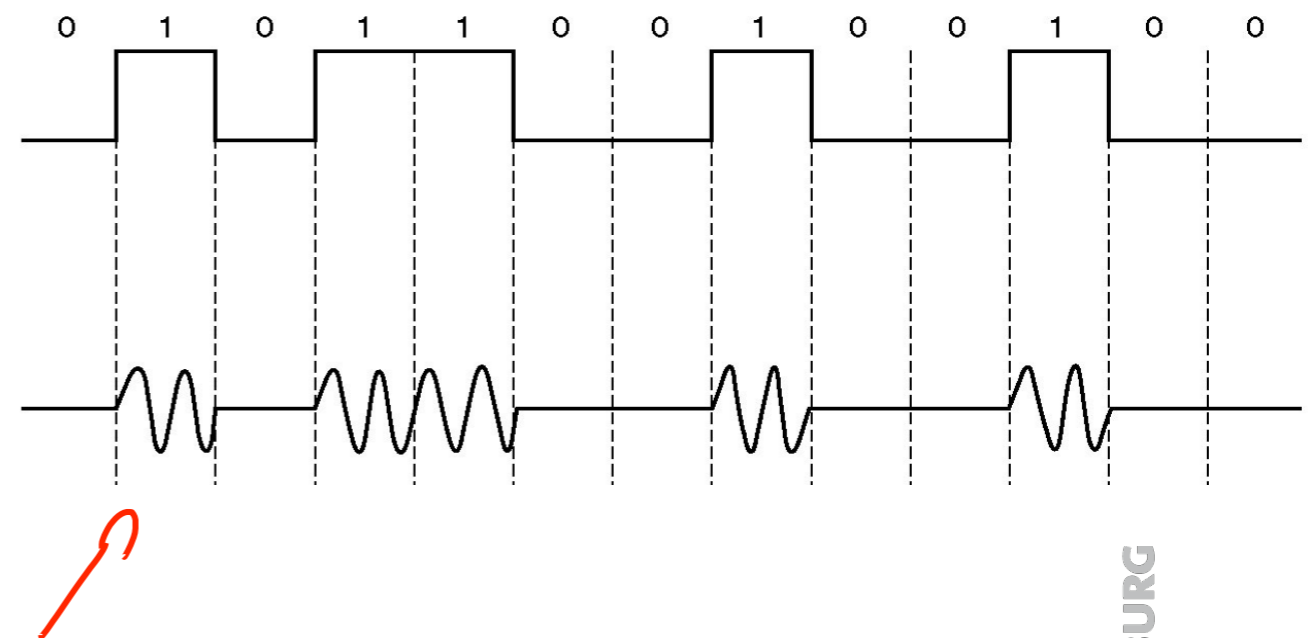
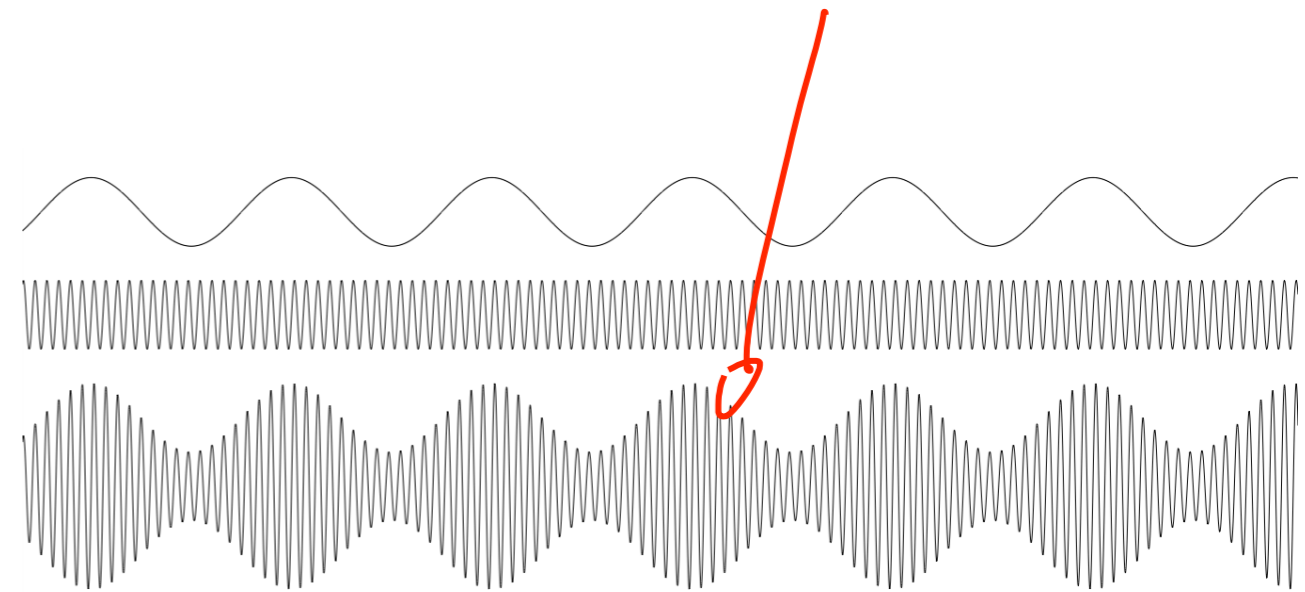
$$s(t) = A \sin(2\pi f t + \phi)$$



- The time-varying signal  $s(t)$  is encoded as the amplitude of a sine curve:

$$f_A(t) = s(t) \sin(2\pi ft + \phi)$$

- Analog Signal
- Digital signal
  - amplitude keying
  - special case: symbols 0 or 1
    - on / off keying



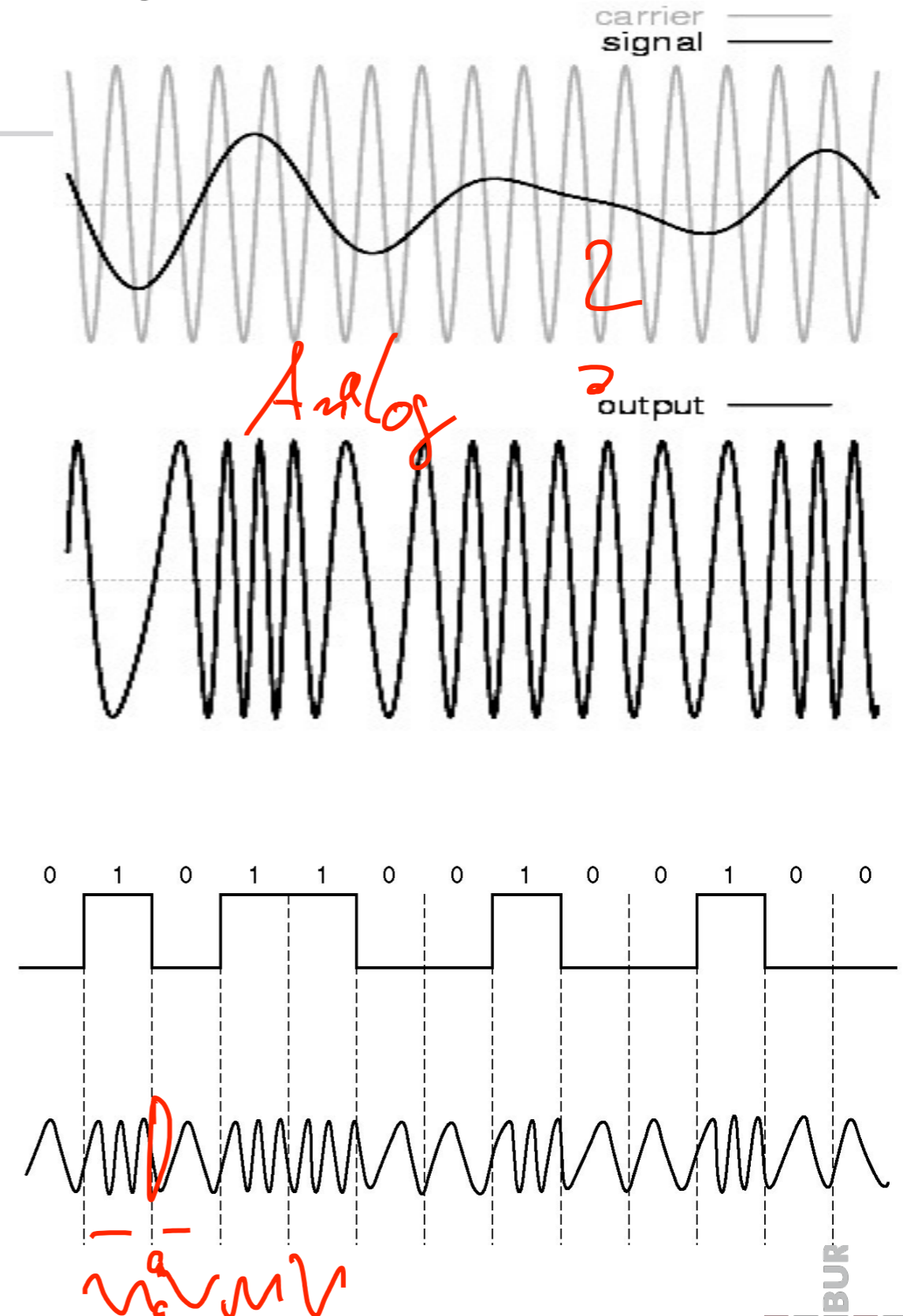


# Frequency Modulation

- The time-varying signal  $s(t)$  is encoded in the frequency of the sine curve:

$$f_F(t) = a \sin(2\pi s(t)t + \phi)$$

- Analog signal
  - Frequency modulation (FM)
  - Continuous function in time
- Digital signal
  - Frequency Shift Keying (FSK)
  - E.g. frequencies as given by symbols

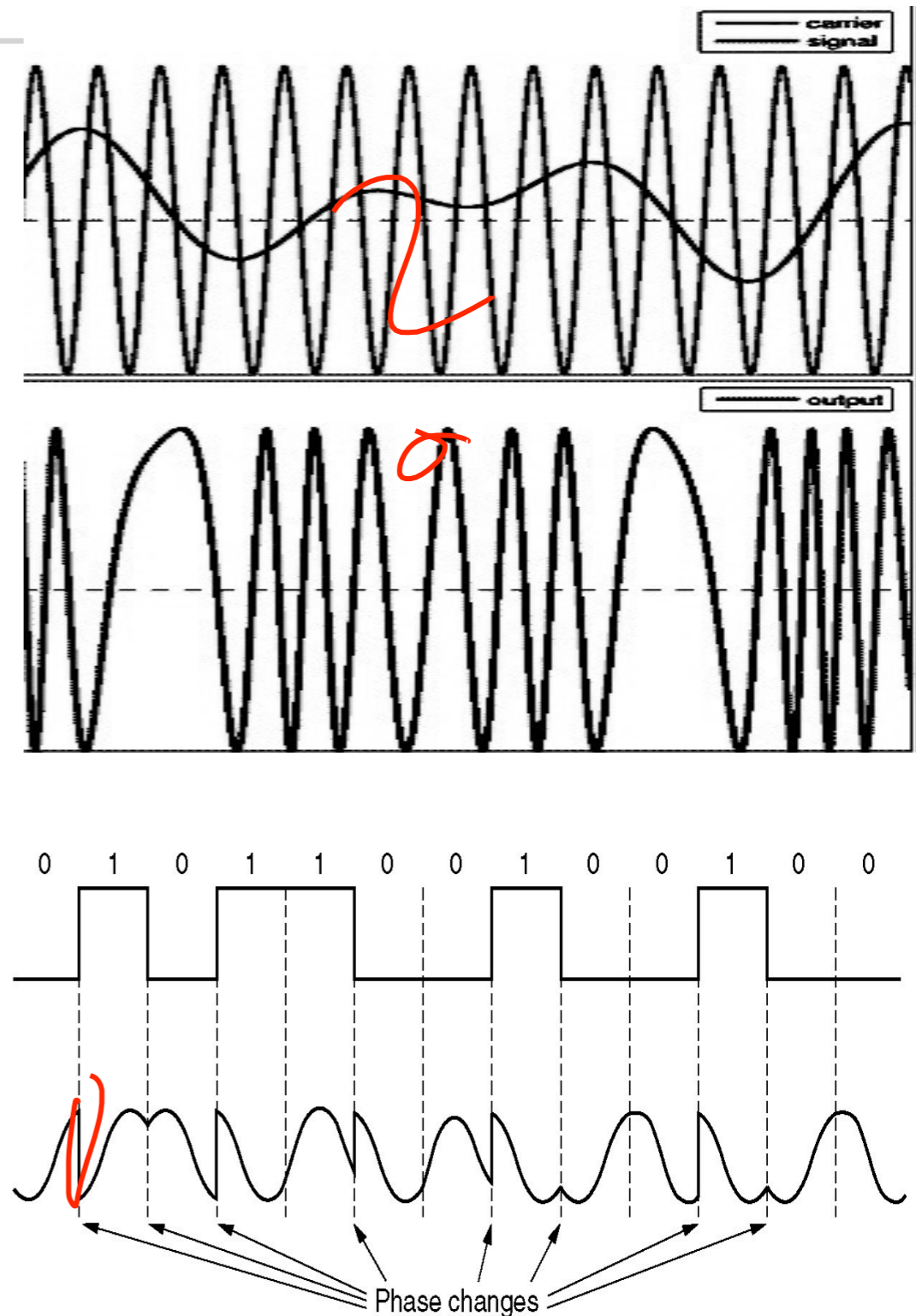


# Phase Modulation

- The time-varying signal  $s(t)$  is encoded in the phase of the sine curve:

$$f_P(t) = a \sin(2\pi ft + s(t))$$

- Analog signal
  - phase modulation (PM)
  - very unfavorable properties
  - es not used
- Digital signal
  - phase-shift keying (PSK)
  - e.g. given by symbols as phases

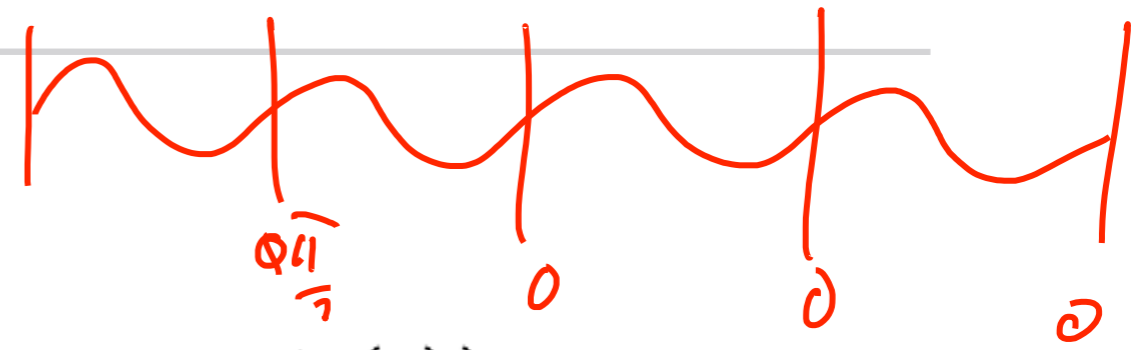


- For a station there are two options
  - digital transmission
    - finite set of discrete signals
    - e.g. finite amount of voltage sizes / voltages
  - analog transmission
    - Infinite (continuous) set of signals
    - E.g. Current or voltage signal corresponding to the wire
- Advantage of digital signals:
  - There is the possibility of receiving inaccuracies to repair and reconstruct the original signal
  - Any errors that occur in the analog transmission may increase further

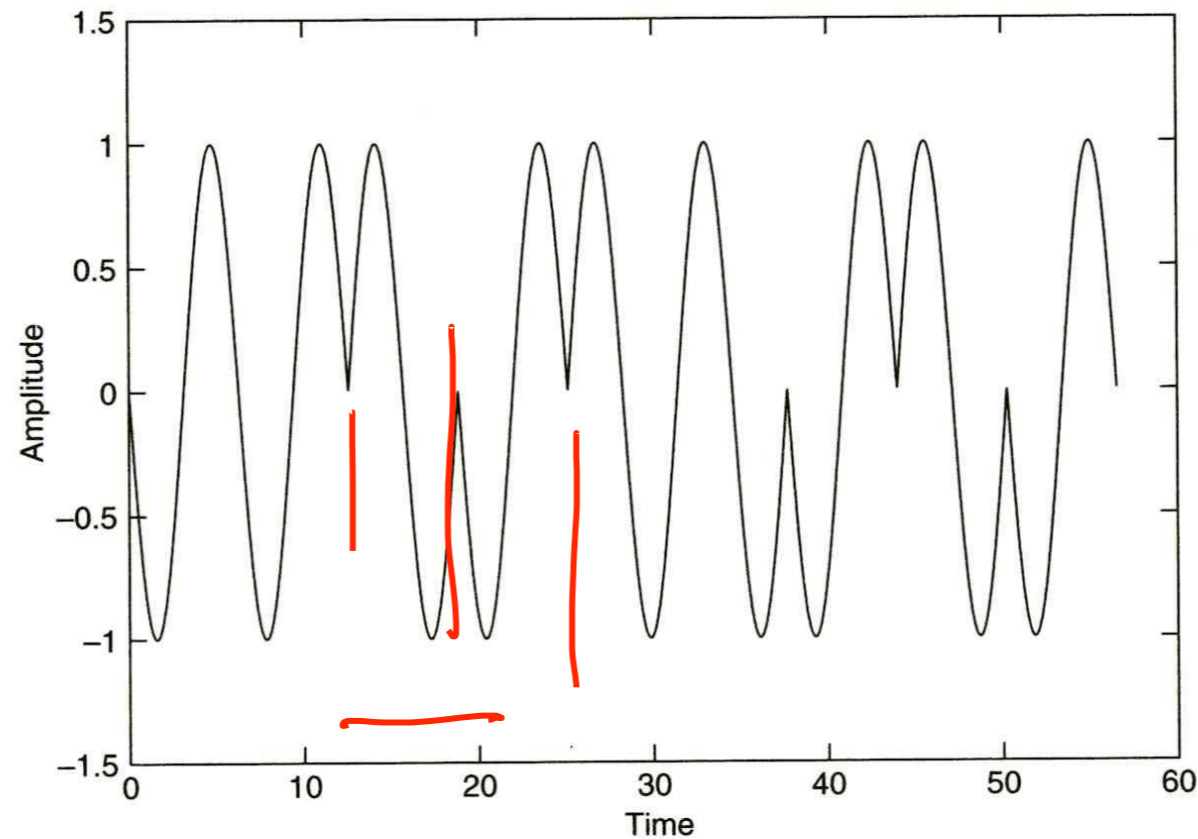
# Phase Shift Keying (PSK)

- For phase signals  $\phi_i(t)$

$$s_i(t) = \sqrt{\frac{2E}{T}} \cdot \sin(\omega_0 t + \phi_i(t))$$



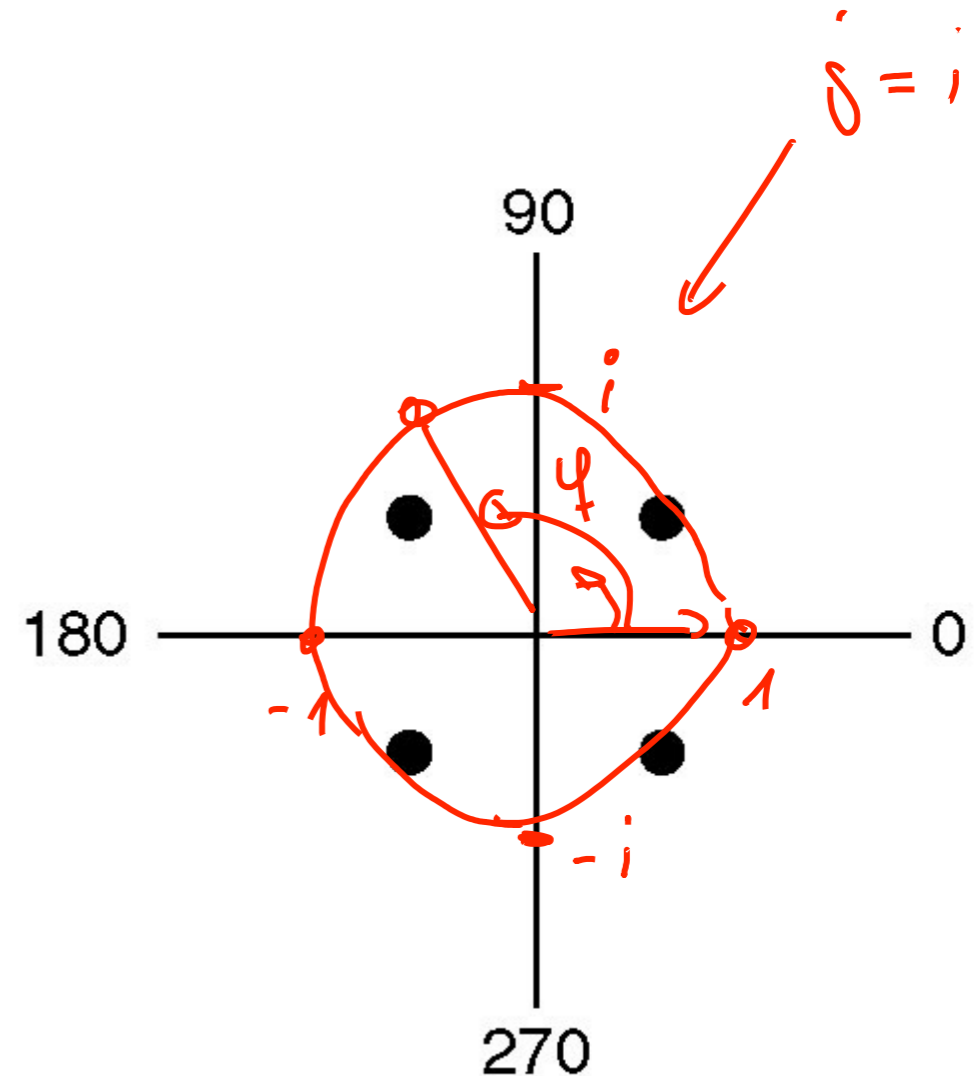
- Example:



*→ Synchronizati.*

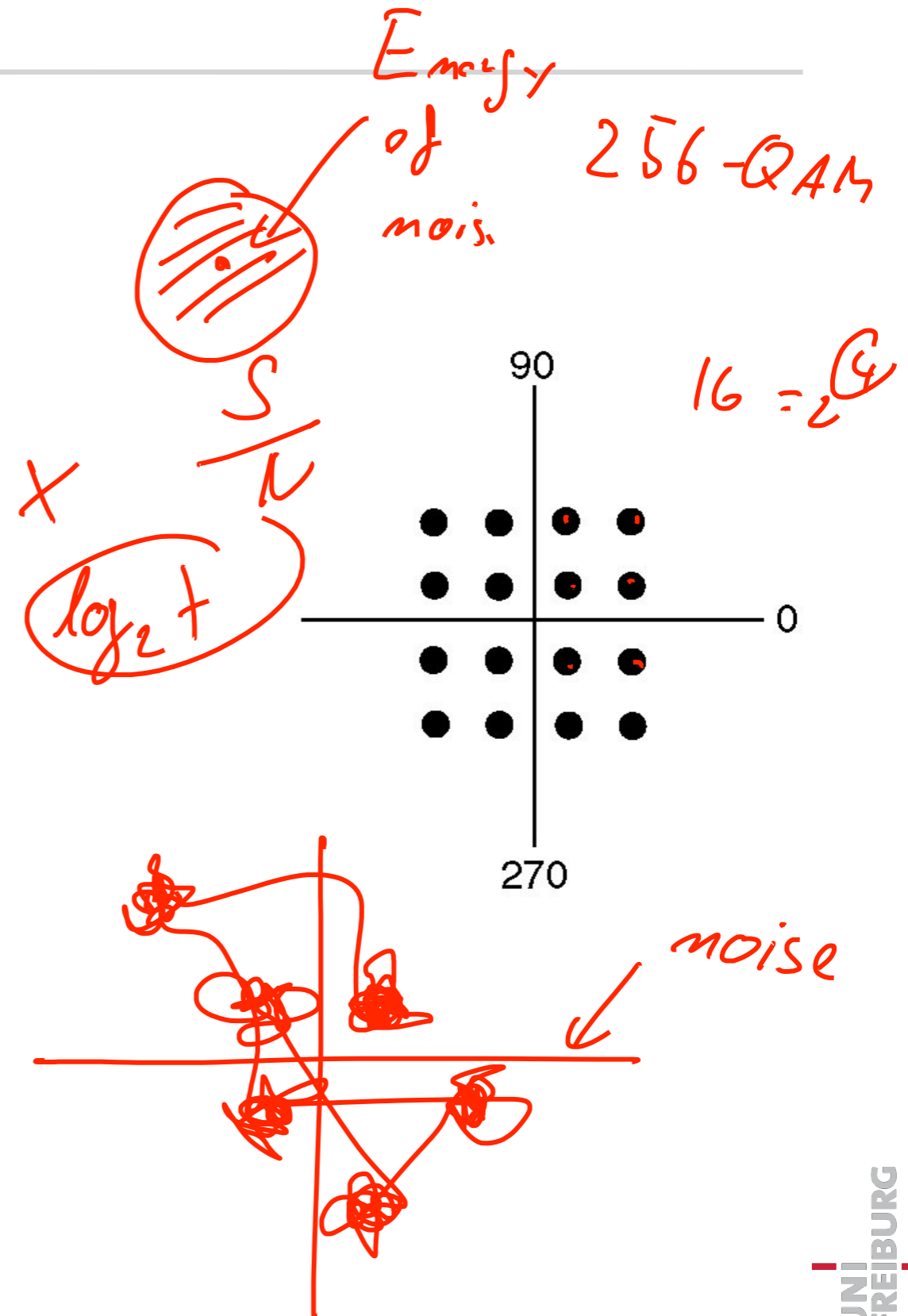
# PSK with Different Symbols

- Phase shifts can be detected by the receiver very well
- Encoding various Symbols very simple
  - Using phase shift e.g.  $\pi / 4$ ,  $3/4\pi$ ,  $5/4\pi$ ,  $7/4\pi$ 
    - rarely: phase shift 0 (because of synchronization)
  - For four symbols, the data rate is twice as large as the symbol rate
- This method is called Quadrature Phase Shift Keying (QPSK)



# Amplitude and Phase Modulation

- Amplitude and phase modulation can be successfully combined
  - Example: 16-QAM (Quadrature Amplitude Modulation)
    - uses 16 different combinations of phases and amplitudes for each symbol
    - Each symbol encodes four bits ( $2^4 = 16$ )
  - The data rate is four times as large as the symbol rate



- Definition

- The band width  $H$  is the maximum frequency in the Fourier decomposition

- Assume

- The maximum frequency of the received signal is  $f = H$  in the Fourier transform
  - (Complete absorption [infinite attenuation] all higher frequencies)
- The number of different symbols used is  $V$
- No other interference, distortion or attenuation of

- Nyquist theorem

- The maximum symbol rate is at most  $2 H$  baud.
- The maximum possible data rate is a bit more than  $2 \log_2 H V / s$ .



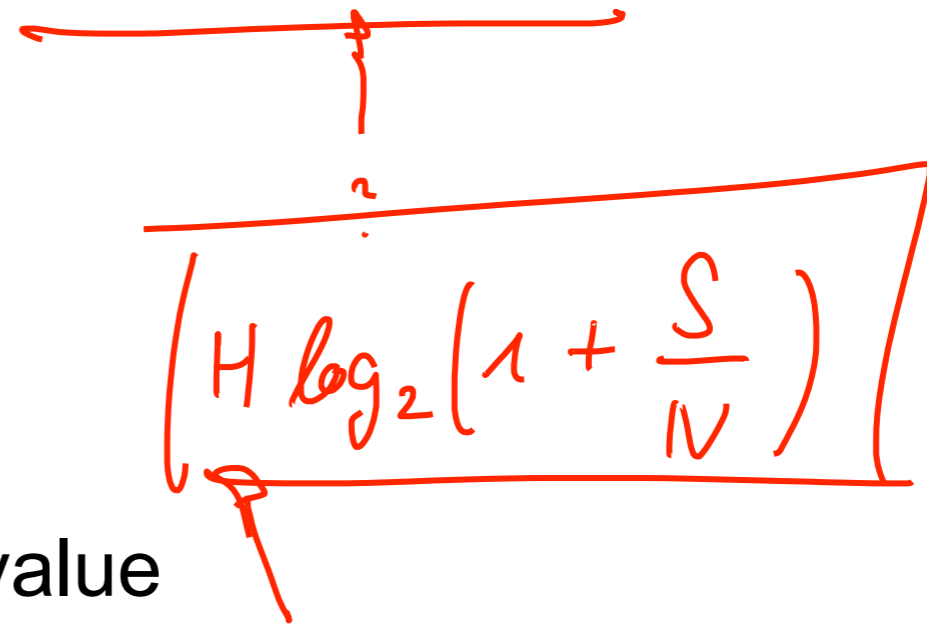
# Do more symbols help?

- Nyquist's theorem states that could theoretically be increased data rate with the number of symbols used
- Discussion:
  - Nyquist's theorem provides a theoretical upper bound and no method of transmission
  - In practice there are limitations in the accuracy
  - Nyquist's theorem does not consider the problem of noise



# The Theorem of Shannon

- Indeed, the influence of the noise is fundamental
  - Consider the relationship between transmission intensity  $S$  to the strength of the noise  $N$
  - The less noise the more signals can be better recognized
- Theorem of Shannon
  - The maximum possible data rate is  $H \log_2(1 + S / N)$  bits/s
    - with bandwidth  $H$
    - Signal strength  $S$
- Attention
  - This is a theoretical upper bound
  - Existing codes do not reach this value

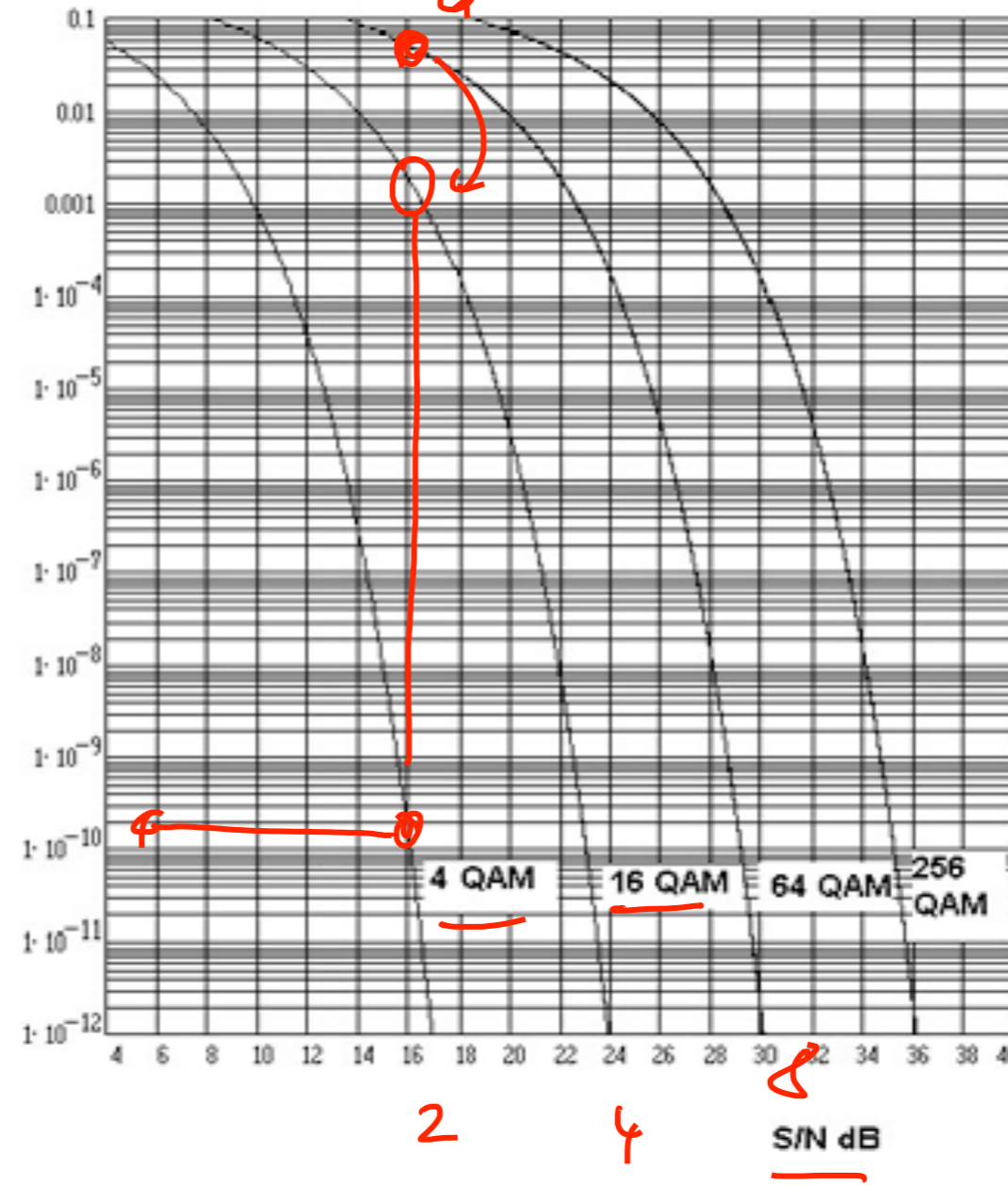


Handwritten red annotations around the formula  $H \log_2(1 + S/N)$ . A horizontal line with a vertical tick mark is drawn above the formula. A vertical line with a horizontal tick mark is drawn to the left of the formula. The formula itself is enclosed in a red rectangular box. A red arrow points to the bottom-left corner of the box.

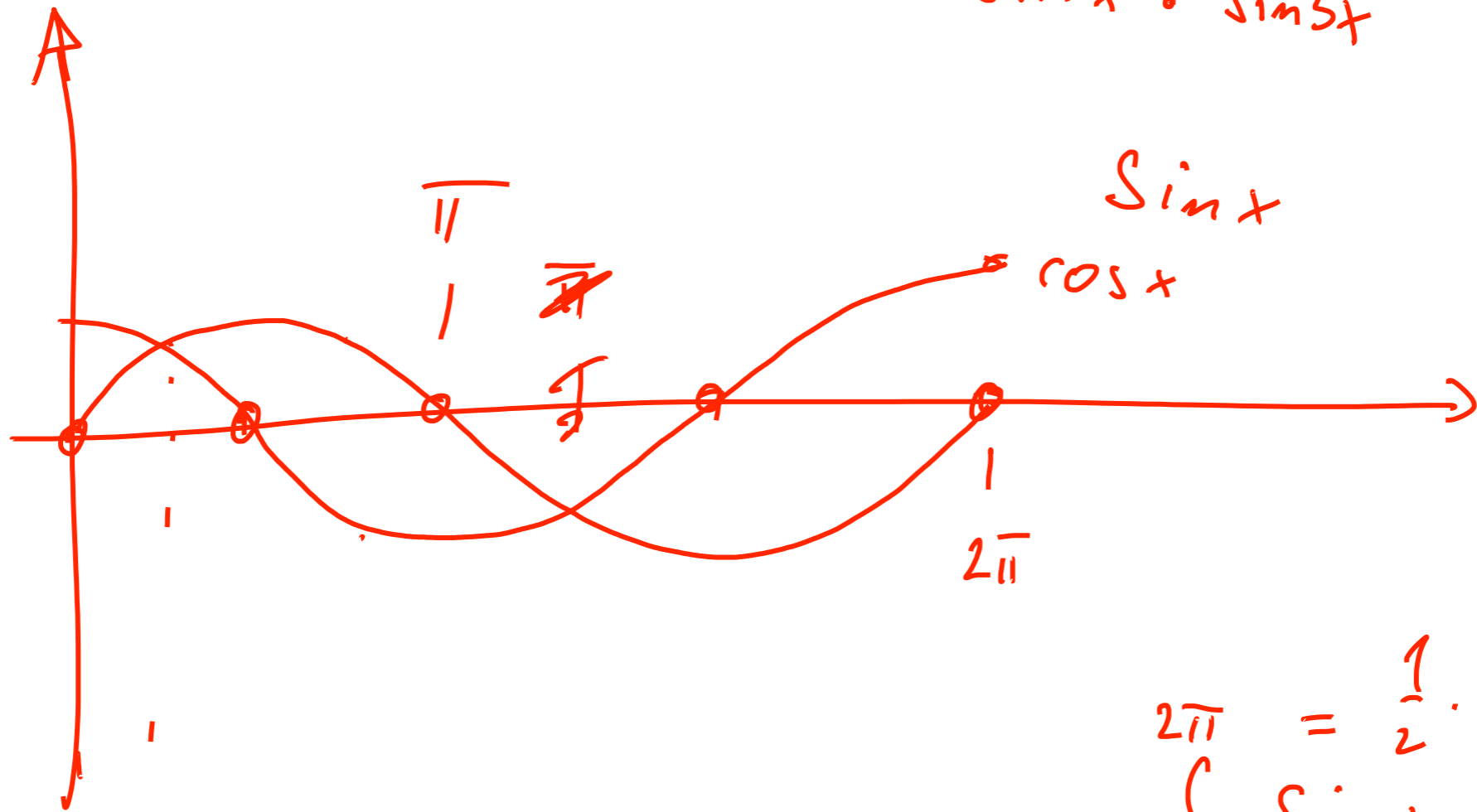
$$\left( H \log_2 \left( 1 + \frac{S}{N} \right) \right)$$

# Bit Error Rate and SINR

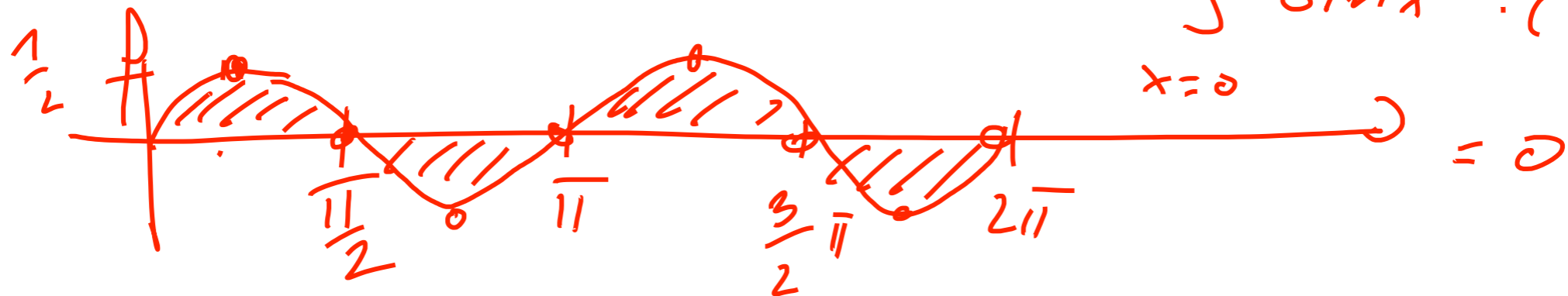
- Higher SINR decreases Bit Error Rate (BER)
  - BER is the rate of faulty received bits
- Depends from the
  - signal strength
  - noise
  - bandwidth
  - encoding
- Relationship of BER and SINR
  - Example: 4 QAM, 16 QAM, 64 QAM, 256 Q. ....

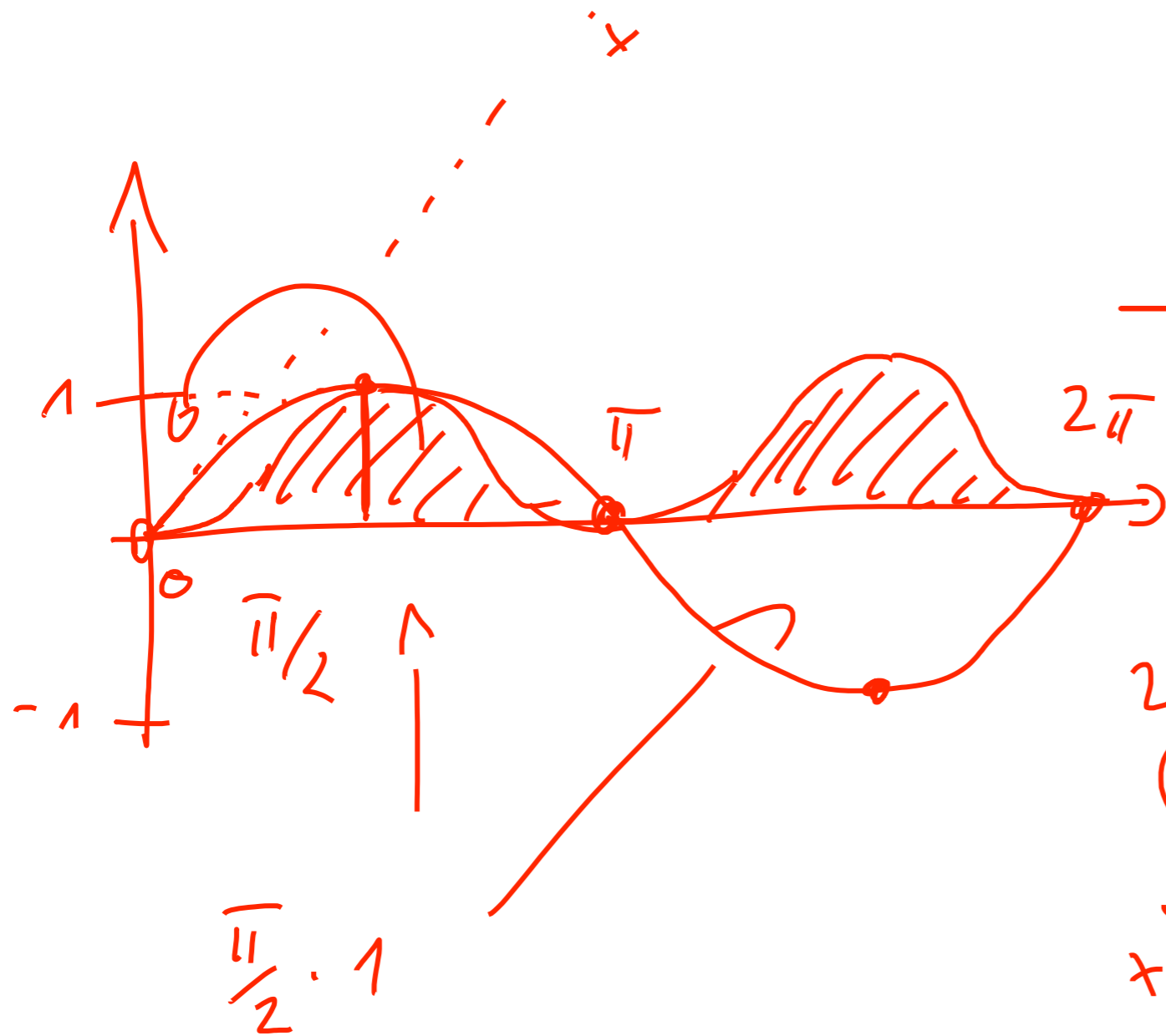


$$\sin x \cdot \sin 3x \quad 0 \cdot 1 = 0$$



$$\int_0^{2\pi} \sin x \cdot \cos x \, dx = \frac{1}{2} \int_0^{2\pi} \sin 2x \, dx$$





$$(\sin x)(\sin x)$$

~~Δ~~

$$= 1 - \frac{1}{2} \cos 2x$$

$$\int_0^{2\pi} \sin x \, dx = 0$$

c c' g' c'' e'' g'' c'''  
 1 2 3 4 5 6 7 8

