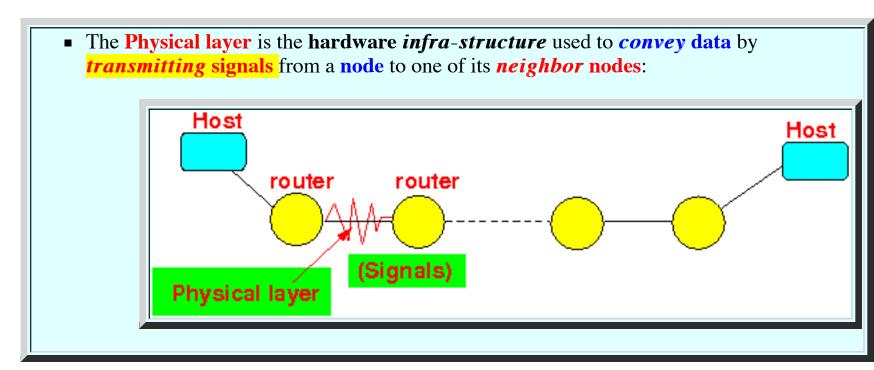
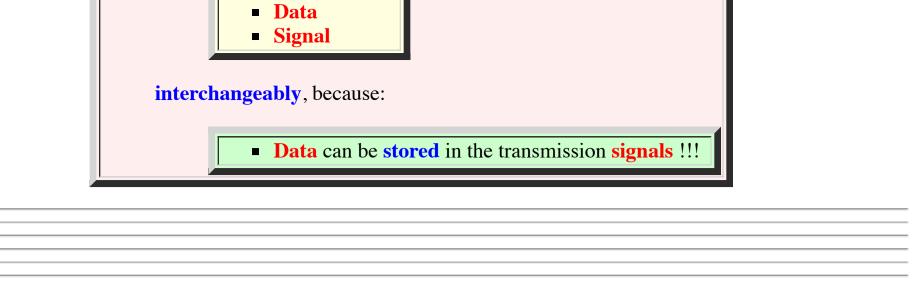
### **Physical Layer**

- The Physical Layer
  - Purpose of the Physical layer:

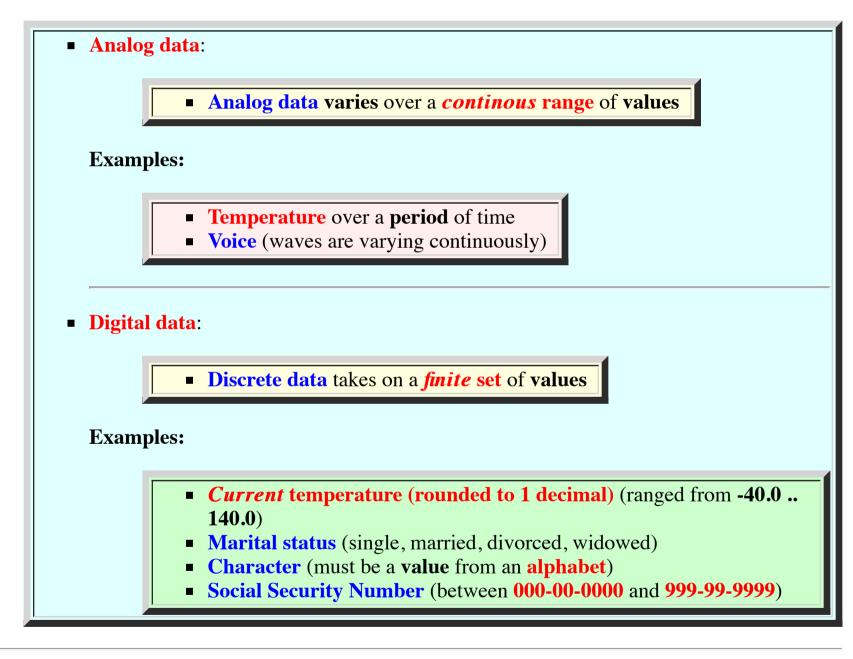


- Important note:
  - The physical layer deals *only* with nodes that can be reached *directly*

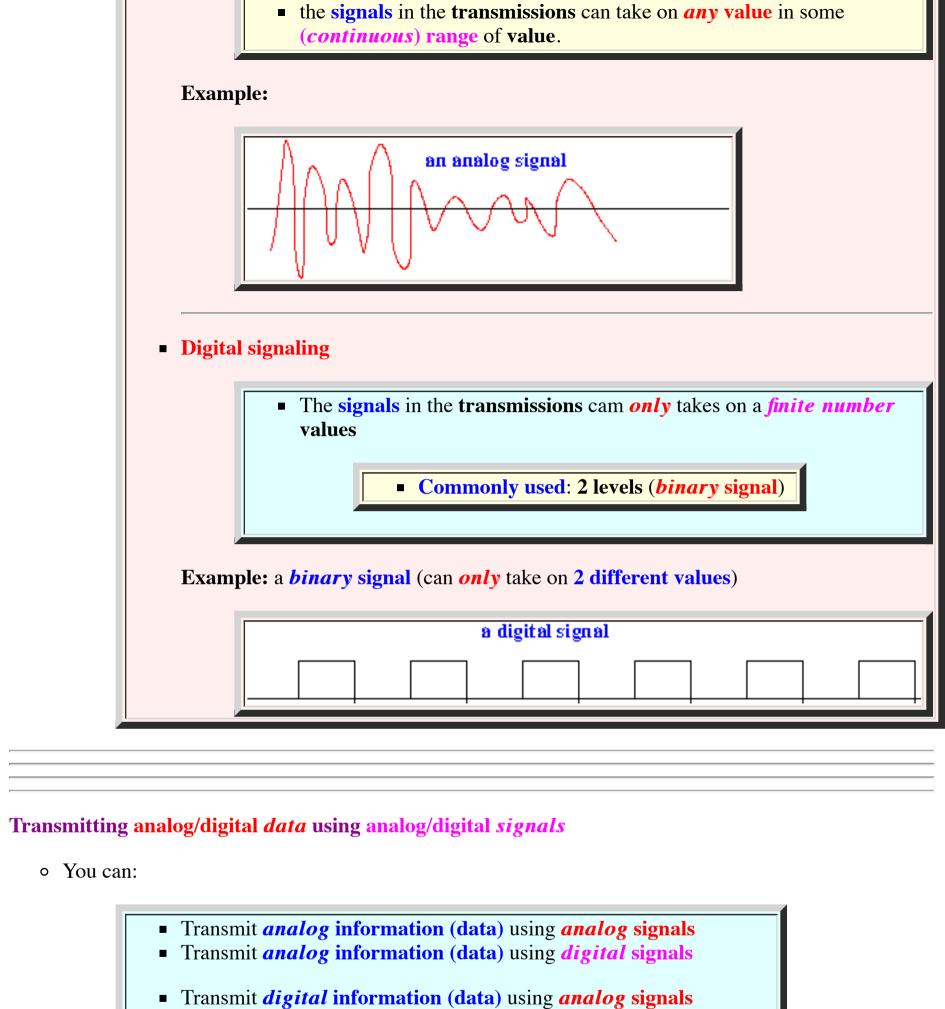
# Information, data and signals • Information and data • Data: ■ Data = A set of *qualitative* or *quantitative* values • E.g.: Dentist, Feb 2, 10:00 AM • Information: ■ Information = the meaning that humans impart on data ■ E.g.: I have a dentist appointment on Feb 2, at 10 AM. **Data** communication • Fact: • We can *only* transmit data.... ■ We do *not* transmit *information* Data and signals • Data transmission: ■ In computer communication, data is transmitted using (electrical) signals • Note: ■ I sometimes use



- Types of data
  - There are 2 types of data:



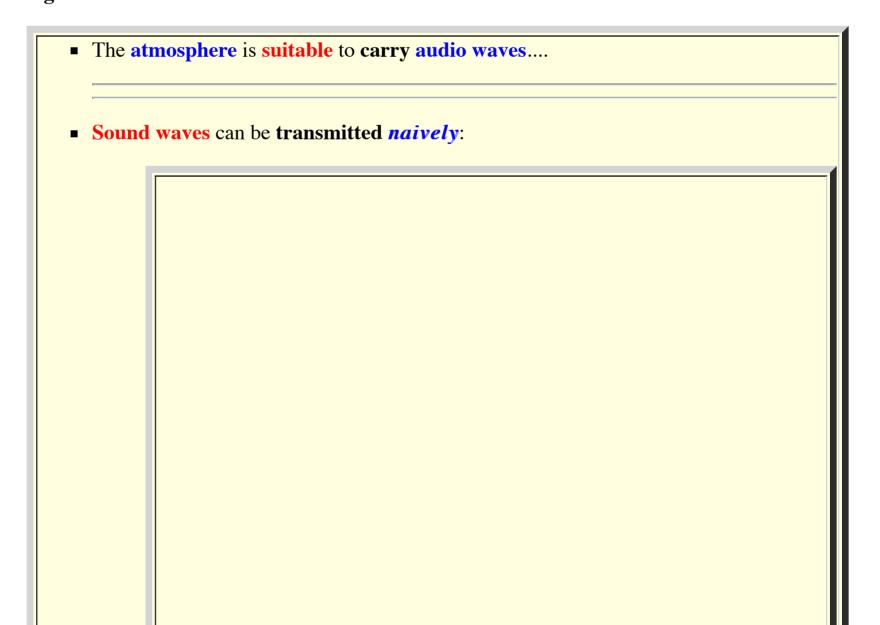
- Types of signaling methods (signals to send data)
  - Data communication can use 2 different signaling methods:
    - Analog signaling



■ Transmit digital information (data) using digital signals

#### Transmitting analog data using analog signals

- Analog data transmitted with analog signals
  - 2 ways to transmit analog data using analog signals:
    - If the transmission medium is suitable (i.e., it can carry the transmitted signals), then:
       If the transmission medium is not suitable (i.e., it does not carry the transmitted signals), then:
       Modulate (super-impose) the signal on a carrier signal
       Transmit the modulated signal
- Example a *naive* transmission
  - "Talking":



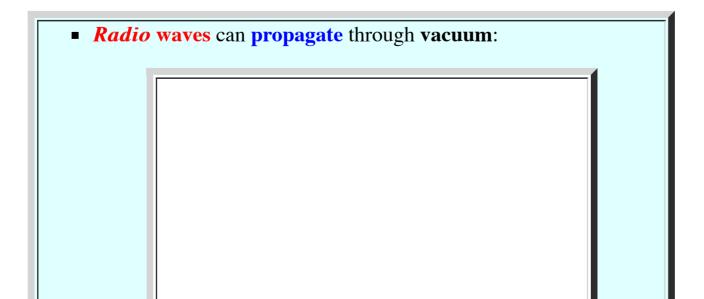


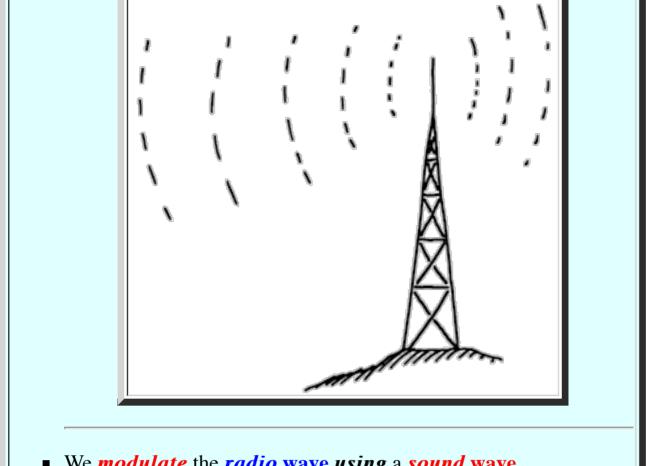
#### • Intro to Modulation

#### • Problem:

- How can astronauts talk on the moon ???
  - The moon does not have an atmosphere
  - **Sound wave** is *not* suitable

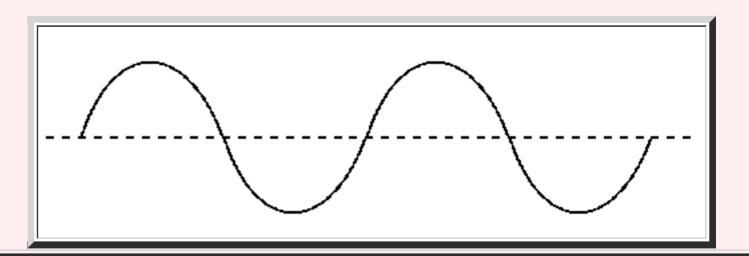
#### • Solution:





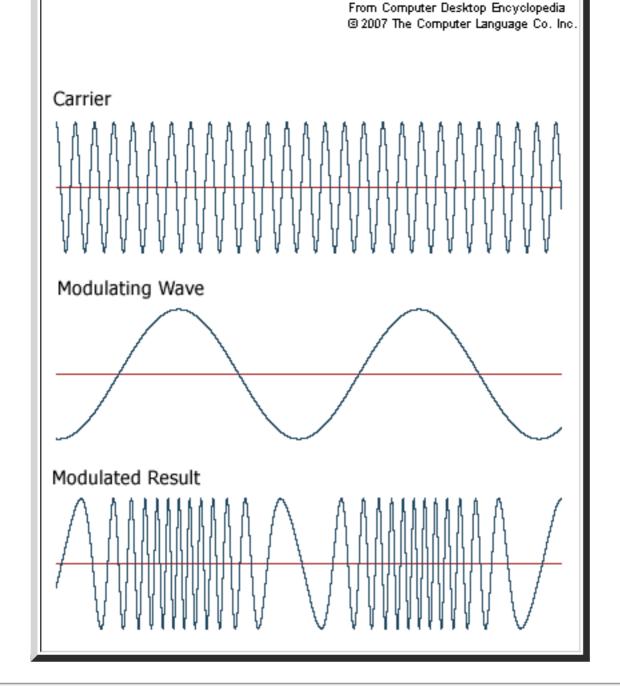
- We *modulate* the *radio* wave using a sound wave
- Then transmit the *modulated* radio wave (through vacuum)....
- What is "Modulation"
  - Carrier wave:
    - Carrier wave = a sine wave of a specific frequence and amplitude that propagates easily through the transmission medium

Example: carrier wave

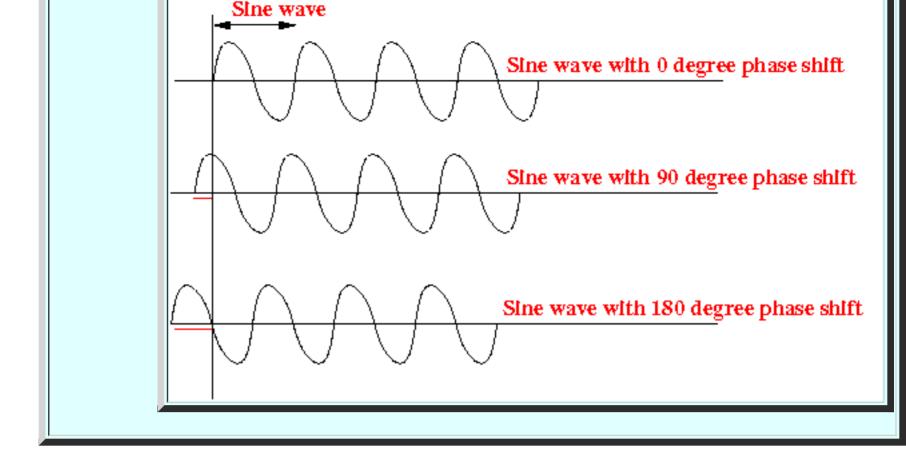


- Modulation:
  - Modulation = *change* a carrier wave using some (input) signal

• What can you change in the carrier wave: ■ The amplitude of the carrier wave (= Amplitude Modulation (AM) The frequency of the carrier wave (= Frequency Modulation (FM) The phase of the carrier wave (= Phase Modulation (PM) **Amplitude Modulation** • Amplitude modulation (AM)): • the amplitude of the carrier wave is changed according to the input signal **Example:** CARRIER SIGNAL • Frequency Modulation (FM) • Frequency modulation (FM)): • the **frequency** of the **carrier wave** is **changed** according to the **input signal Example:** 



- Phase Modulation (PM)
  - *Phase* modulation (PM)):
    - the phase of the carrier wave is changed according to the input signal
  - The **phase** of a **wave**:
    - The phase of a sine wave is the *amount* of shift:



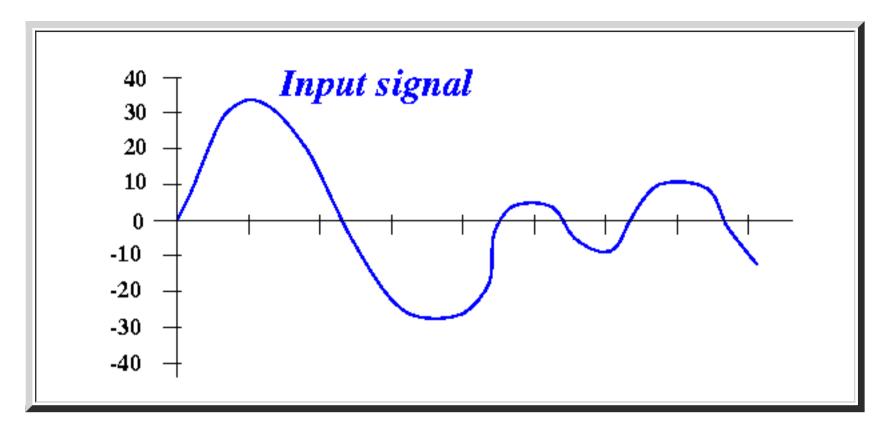
- In PM, you change the instantaneous phase of the carrier signal.
  - This is **very hard** to **detect**, even for **electronics** let alone **humans** !!!

I will **not try** to **draw** a **continuous phase shifted** signal.....

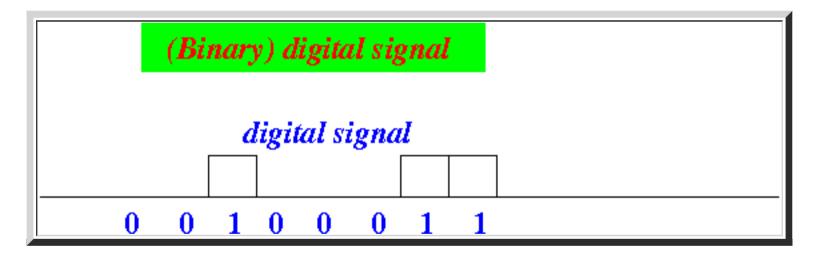
It's **too** hard:)

## Converting analog data (signals) to digital data (signals)

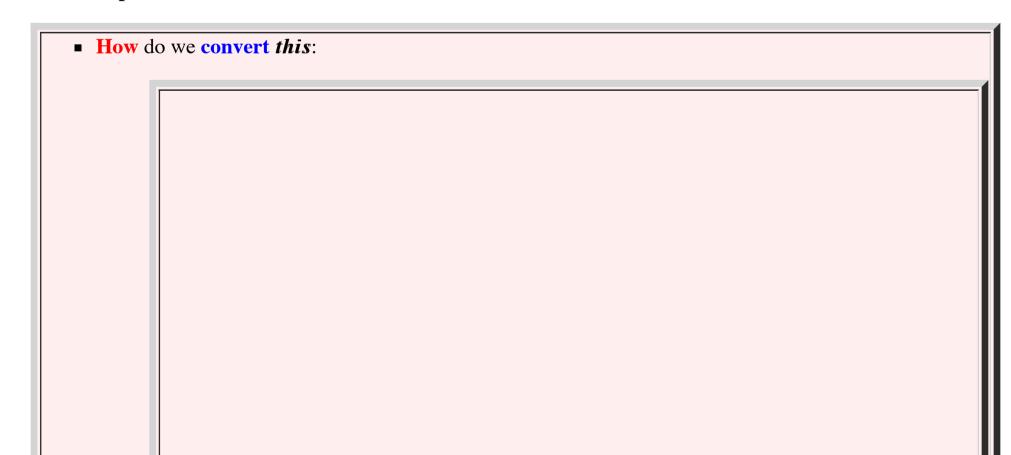
- Transmitting analog data using digital signals
  - Analog (input) data:

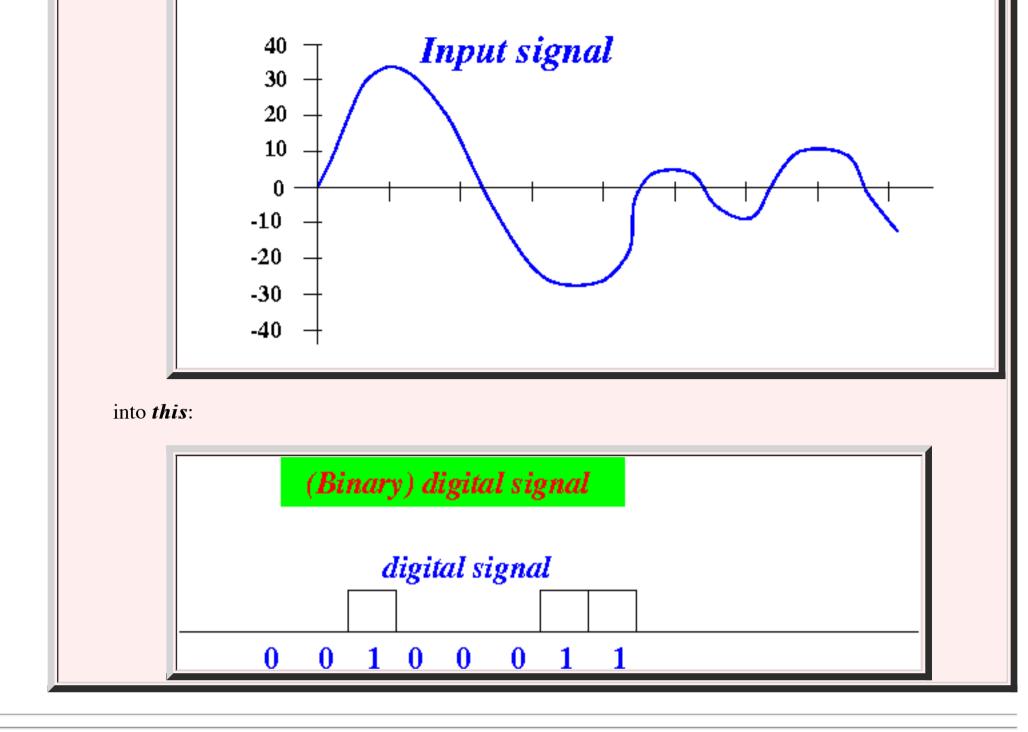


• Digital signals:



• Problem Description:



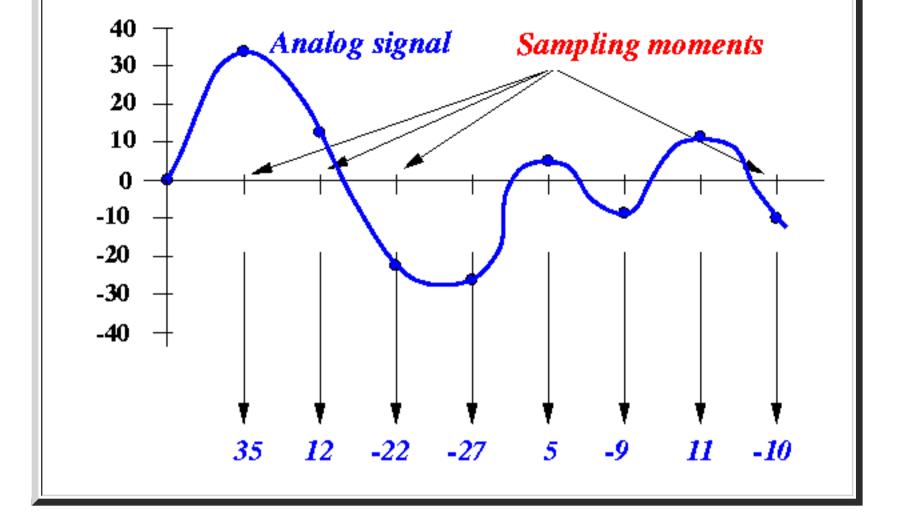


#### • Sampling

- Sampling:
  - Measure (and record) a continuous signal at regular intervals

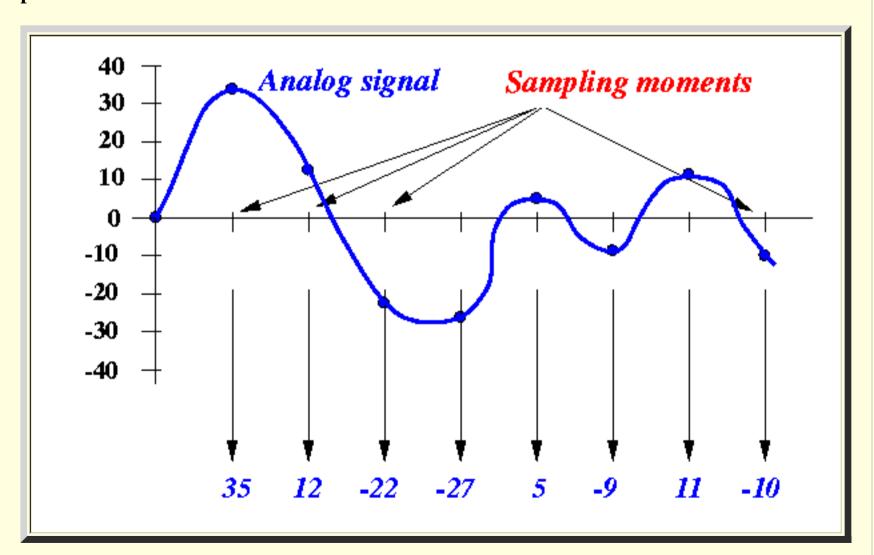
#### **Example:**



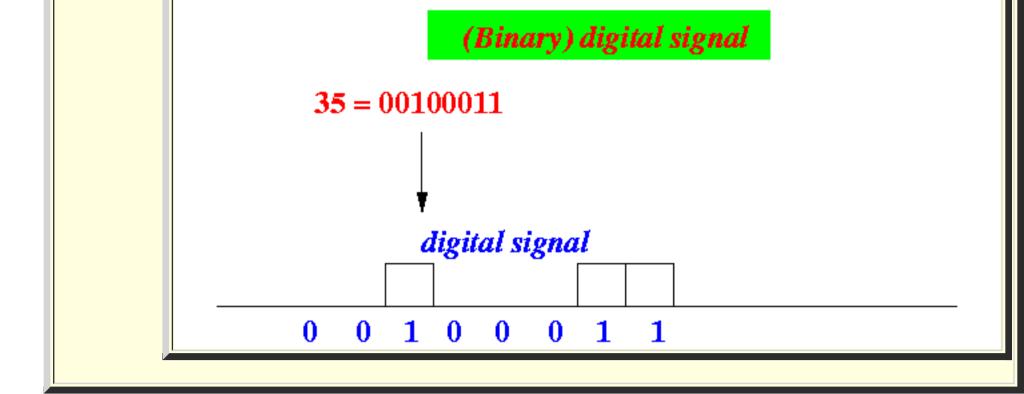


- Converting analog data (signal) to digital data (signal)
  - How to transmit analog data using discrete signals
    - 1. The analog data is first sampled into serie of numerical values

#### **Example:**



2. Each of the *numerical* value is **encoded** into a **binary number** and transmitted using (**binary**) **digital** signals:



#### • A/D converters

- A/D converter (chip):
  - A/D converter = the device (chip) that converts (samples) analog signals to digital signals
    - Wikipedia page: click here

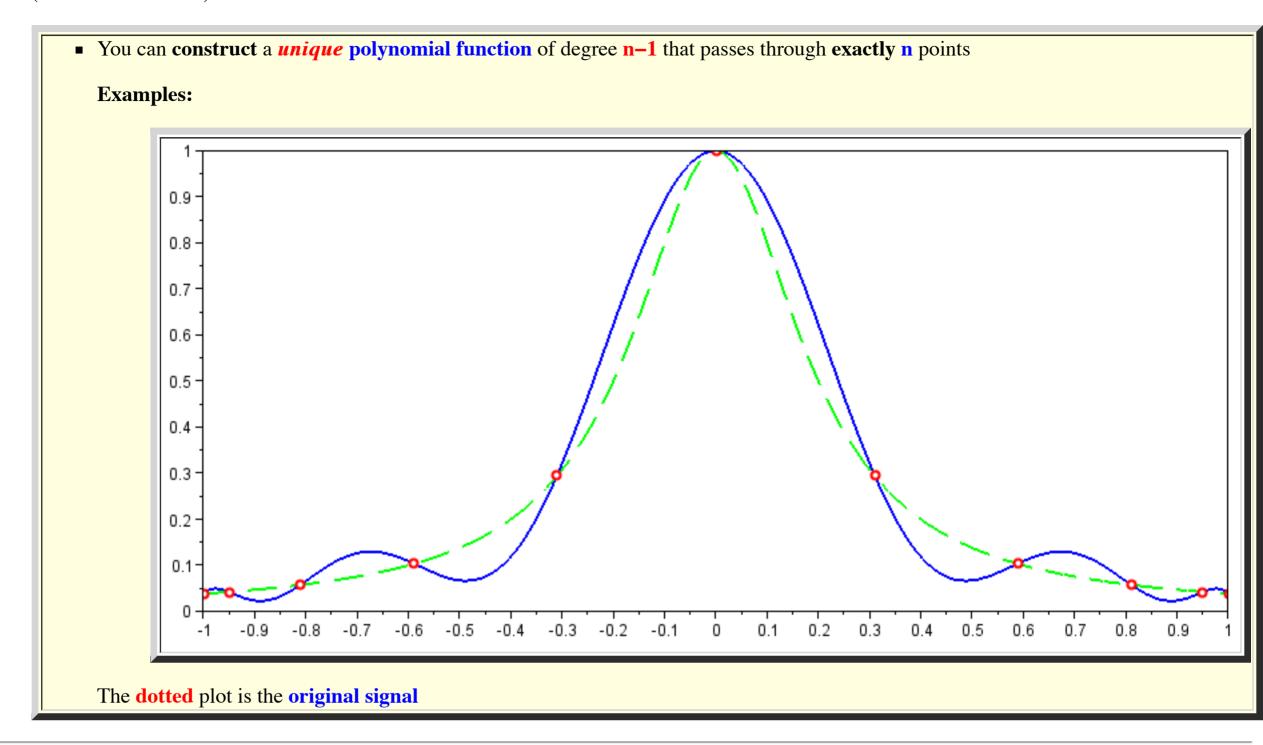
You can specify a sampling rate:

- The more *frequent* you sample the input signal, the better accurate the approximation !!!
- Caveat:
- If you sample below a certain threshold, you may not be able to reconstruct the original signal using the sampled values !!!

First, let's look at **how** you **re-construct** the **original signal** from the **samples** 

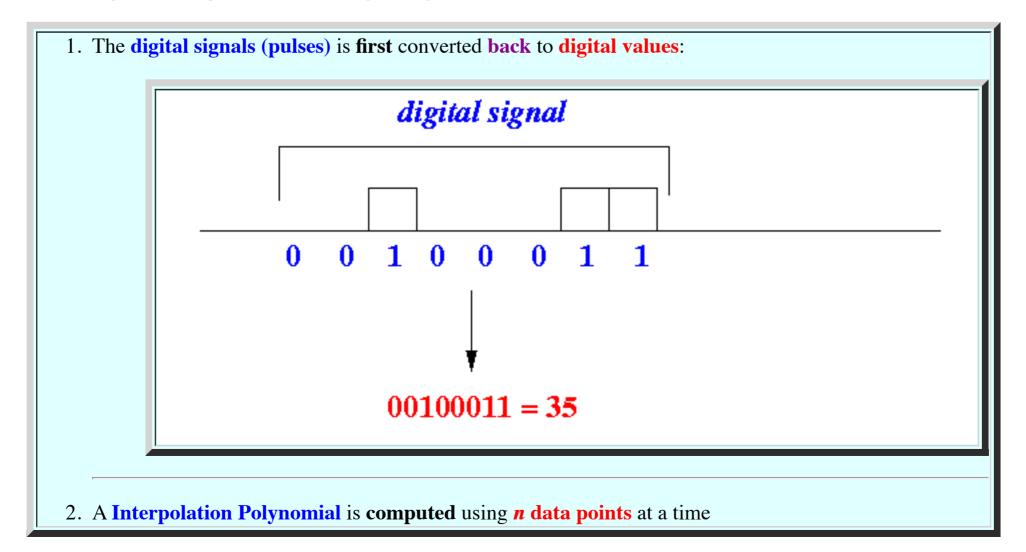
# Re-constructing the *analog* data (signals)

- Re-constructing the analog data from the samples: Lagrange inter-polation
  - Fact: (from Mathematics)



- Langrange Polynomial:
  - Langrange Polynomial = the polynomial function (of degree n-1) that you can construct when given exactly n data points

    See Wikipedia for more details: click here
- Re-constructing the analog data from the digital signal:

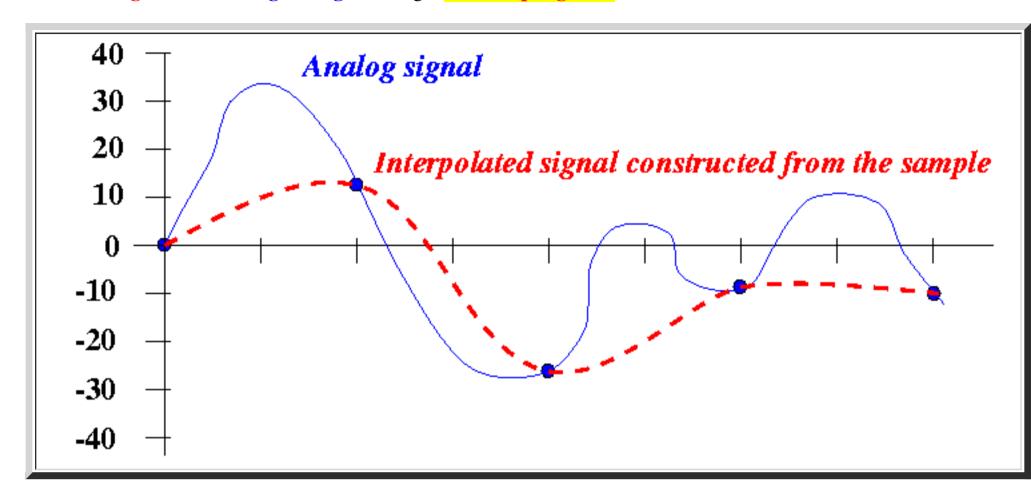


- Here's a nice Language Polynomial applet that you can play with: click here
- I downloaded a less nicer one: <u>click here</u>

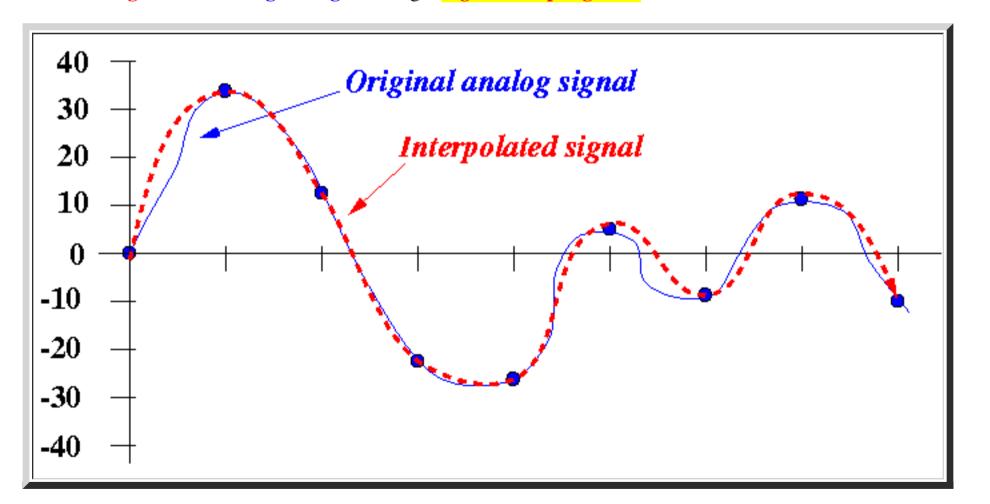
Check out the /home/cs455000/demo/Langrange directory

# Sampling rate and accuracy --- the Nyquist rate

- Sampling rate and accuracy of the reconstructed analog signal
  - Re-constructed signal and the original signal using a low sampling rate:



• Re-constructed signal and the original signal using a higher sampling rate:

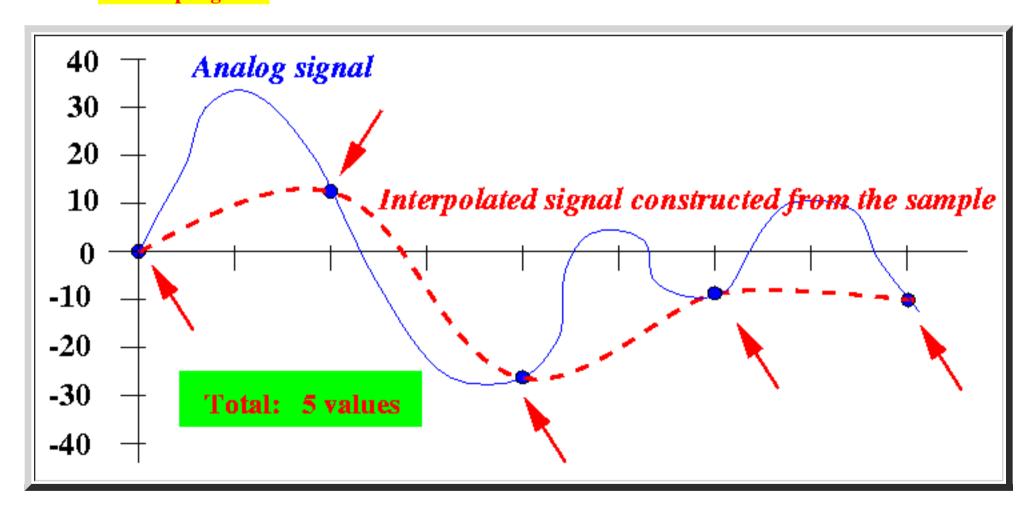


- Conclussion:
  - The *faster* (more frequent) you sample, the *more* accurate the approximation
- \$64,000 question:

■ **Should** we **sample** as **fast** as **possible** ????

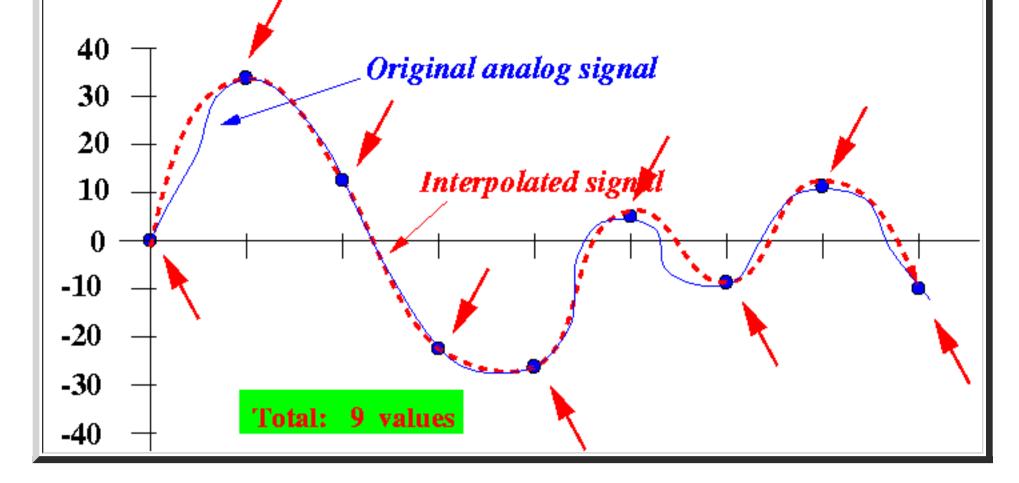
**No** !!!! The **reason** is **cost** !!

- Transmission rate
  - Transmission rate:
    - Transmission rate = the amount of data (in bits) transmitted per time unit (sec)
  - Fact:
- We should try to keep the transmission rate as low as possible
   (More data sent = higher cost)
- Sampling rate and transmission rate
  - Data rate at a *low* sampling rate:



**Conclusion:** 

- We only need to sent 5 values when we sample at the lower sampling rate
- Data rate at a higher sampling rate:



#### **Conclusion:**

- We must sent 9 values when we sample at the higher sampling rate !!!!
- Therefore:
  - The *faster* (more frequent) you sample, the *higher* the (transmission) data rate
- What is the **best** sampling rate ???
  - High sampling rate:
    - Good accuracy
    - *High* transmission data rate (costly)
  - Low sampling rate:
    - *Poor* accuracy
    - Low transmission data rate (cost-efficient)
  - Goal: (what we want)
    - At which data rate can we achieve good enough accuracy ????
- The Nyquist Sampling Rate
  - The Nyquist (sampling) rate:

Nyquist (sampling) rate = the minimum sampling rate required to avoid aliasing when sampling a continuous signal.
 See: Wikipedia

Aliasing:

• an effect that causes different signals to become indistinguishable when sampled.

See: Wikipedia

- In plain English:
  - Nyquist rate = the lowest possible sampling rate that permits an accurate reconstruction of a input signal using samples
- Disclaimer:
  - We will **not derive** the **Nyquist rate** :-)...
    because it requires a **lot** of **background knowledge** and is **beyond the scope** of this course...
- I will **only state** the **result** from **Signal Processing**:

```
Nyquist Rate = 2 × B
where:
    B = highest frequency found in the (analog) input signal
```

#### Application of the Nyquist Rate: digital CD quality audio

- Nyquist rate to obtain digital CD quality audio
  - Problem description:
    - We want to store music (which is analog data) as a computer file (which is digital data)
  - Question:
- How fast do you need to sample to reconstruct audible audio waves completely accurately?
- Pre-requisite: a fact from Biology
  - Human ear can hear sound waves with frequency ≤ 20,000 Hz
- The Nyquist sampling rate for human audible signal:

```
Nyquist Sampling Rate = 2 × MaxFrequency (MaxFreq = 20,000 Hz)
= 2 × 20000 Hz
= 40000 Hz

(Meaning: 40000 samples per second)
```

Note:

```
1 Hz (Hertz) = 1 event (sample) per second
```

- Sampling rate used by the music industry
  - Digital music:
    - **Digital music** = **audio** (**music**) stored in **digital format**
    - Use **sampling** to **convert audio** to **digital** !!!
  - Sampling rate used by the music industry:
    - The sample rate used to obtain *CD quality* audio is: 44100 Hz

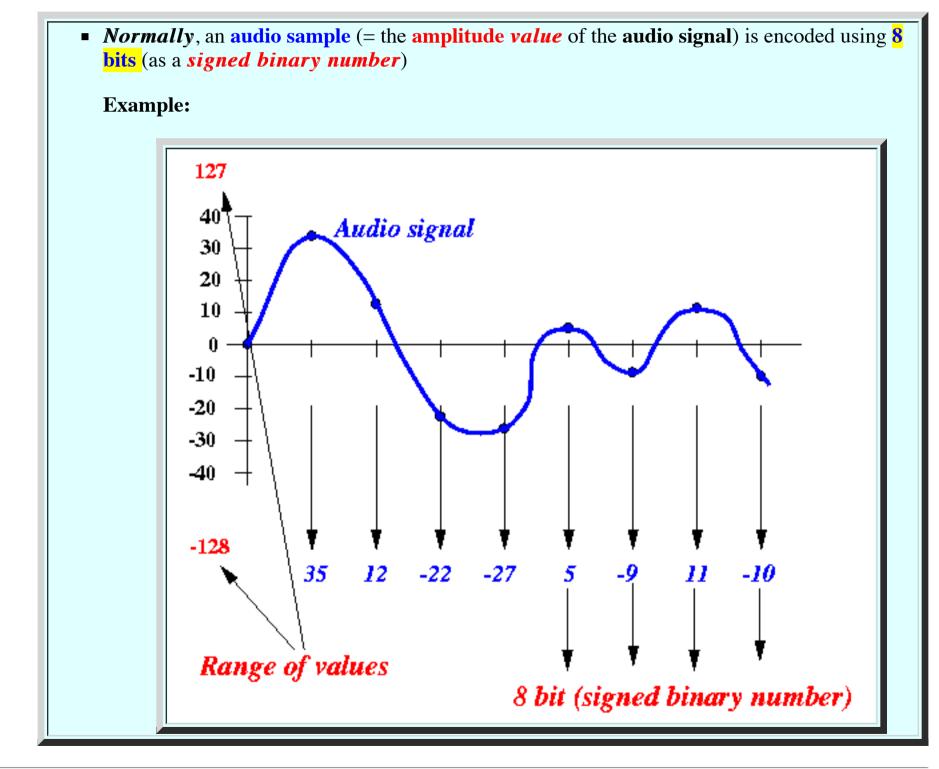
(This rate is **higher** than the **Nyquist** rate of **40,000 Hz**)

■ The reason is historical:

■ The sampling rate of 44.1 kHz was inherited from recording method they already used in video cassettes

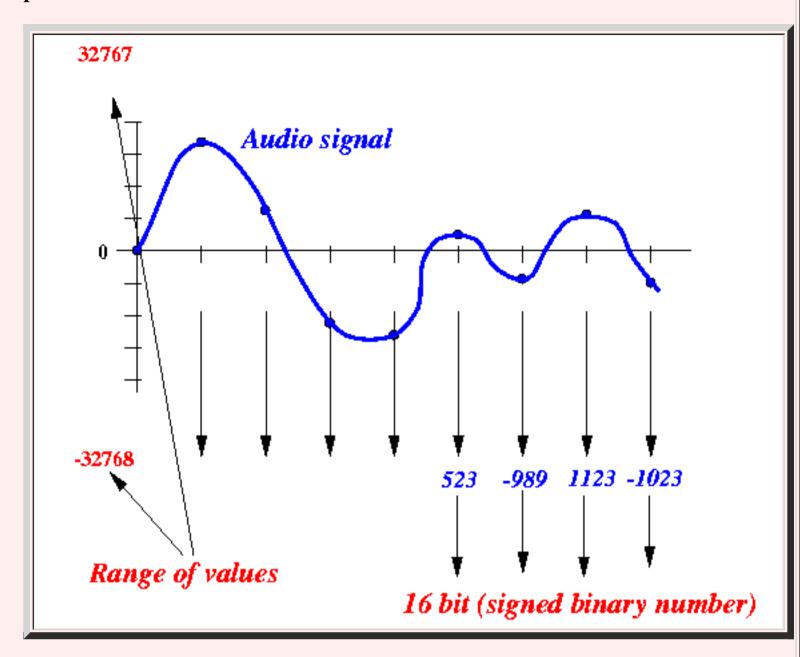
Reference: click here

- Encoding CD quality audio
  - Encoding standard quality audio:



- Encoding **CD quality** audio:
  - In CD quality audio, an audio sample is encoded using 16 bits (signed binary number)

#### **Example:**



(Result: *highly* accurate representation of the **original audio signal**, but will require *more* storage)

- Data rate of CD-quality audio
  - Data rate of mono CD quality audio:

```
Sample Rate = 44,100 Hz (i.e., we sample 44,100 times in 1 sec)

====> 44,100 samples in 1 sec

1 sample is encoded using 16 bits

====> 44,100 × 16 = 705,600 bits in 1 sec

Data rate of (mono) CD quality audio = 705,600 bits/sec
```

• Data *rate* of *mono* CD quality audio:

	====> 2 × 705,600 bits = 1,411,200 bits/sec	
l		

#### **Application of the Nyquist Rate: Digital Telephone**

- Digital telephone
  - Telephone quality audio:
    - Human ear can hear frequencies: 20 20,000 Hz....
       But:
       You do not need the full range to understand what someone is saying...
       Understandble conversions can be carried using the following range of frequencies:
       300Hz 3400 Hz
- Nyquist rate for "telephone" quality audio
  - Maximum frequency of "telephone" quality audio:
    - 3400 Hz !!!
  - Nyquist sampling rate for "telephone" quality audio:

```
Nyquist Sampling Rate (telephone quality) = 2 × MaxFrequency
= 2 × 3400 Hz
= 6800 Hz
i.e.: Minimum sampling rate = 6800 times per second.
```

- Sampling rate used by telephone companies
  - Sampling rate of (digital) telephone:

- Today's (digital) telephone samples the voice at:
   8000 Hz
   I.e.: 8000 samples in 1 sec or 1 sample in 1/8000 sec
- Encoding "telephone" quality audio:
  - Each sample is encoded using signed binary value of 8 bits
- Data rate of "telephone" quality audio
  - Data rate of *telephone quality* audio:

```
Sampling Rate = 8000 Hz

====> 8000 samples in 1 sec

1 sample outputs 8 bits

====> 8000 × 8 bits = 64000 bits in 1 sec

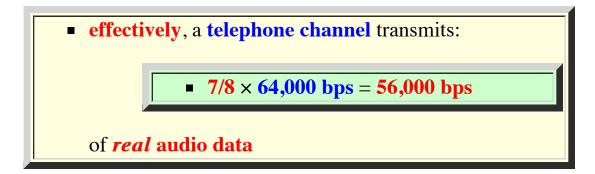
Data rate of telephone quality audio = 64,000 bps
```

- Digital telephone
  - Fact:
- The **telephone company** reserves a **64,000 bps communication channel** for each **phone conversation**

#### **However:**

- The telephone company "steals" one bit from every byte (8 bits) for administration purposes
  - E.g., send the caller ID to the phone

• Effective data rate of a telephone channel:



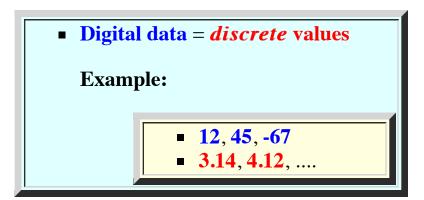
- A gadget from the past
  - The equipment that allows you to transmit data over a telephone connection is:
    - a modem (modulator/de-modulator)
  - Due to the channel capacity of a telephone connection (see above !!!), the modem speed is limitted to 56 kbps !!!



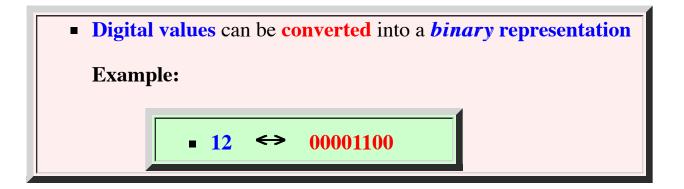
**Modem** is a thing of the **past** because **telecom companies** now sells **Data** Services to their customers.

# Encoding/decoding digital data using digital signals

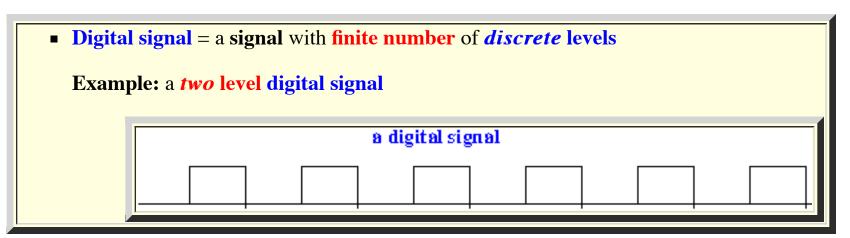
- Digital data
  - Digital data:



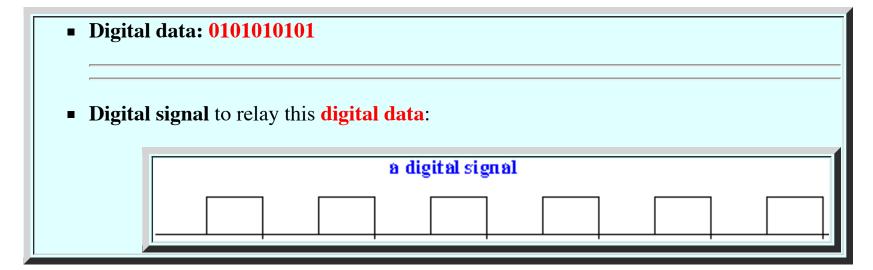
• Fact:



- Digital signal
  - Digital signal:



- Digital data transmission with digital signals
  - Transmitting *digital* data using *digital* signal:

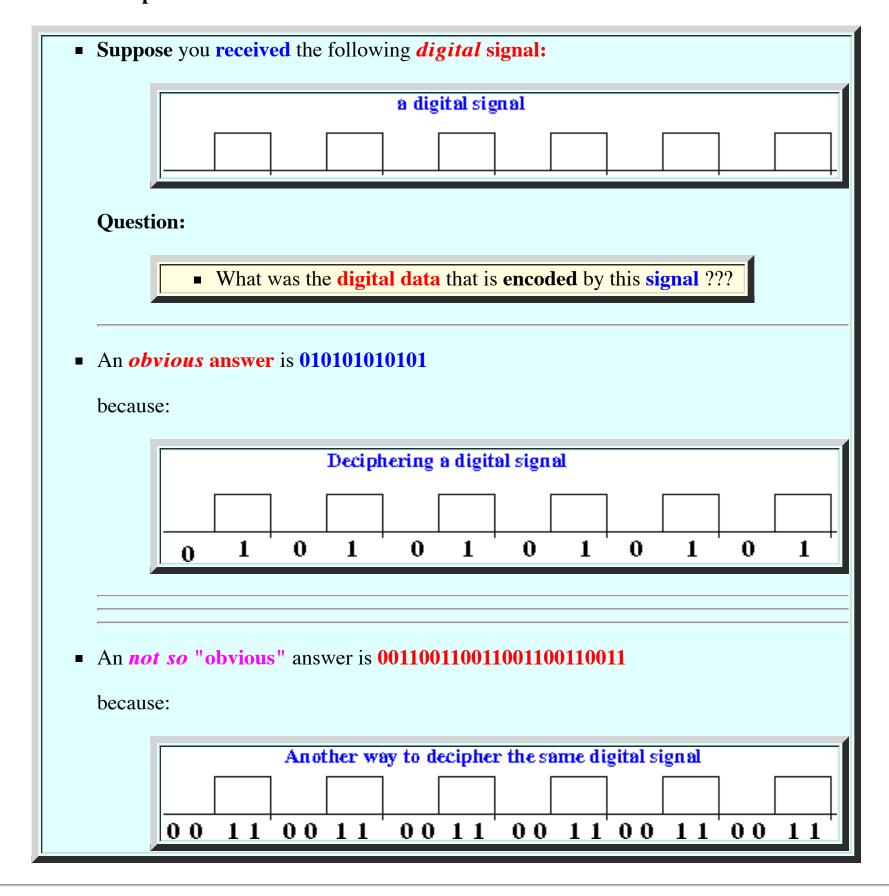


• The missing ingredient in digital transmission

**\$64,000** questions:

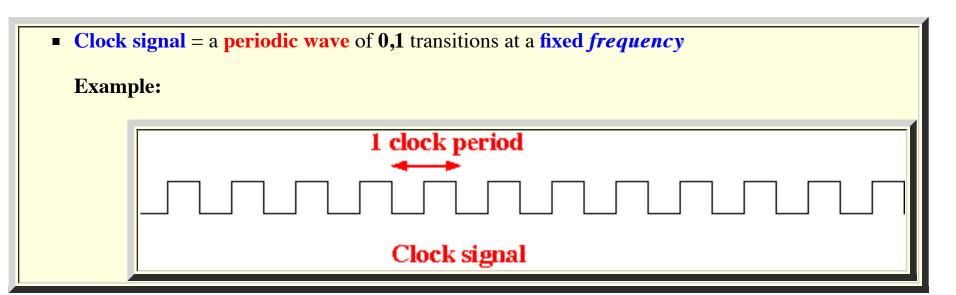
■ Suppose you receive this digital signal:		
a digital signal		
Question:		
■ What was the data transmitted ???		
Obvious answer:		
0101010101		
Is this the <i>only</i> answer ?????		

- Decoding a digital signal
  - *Illustrative* example:

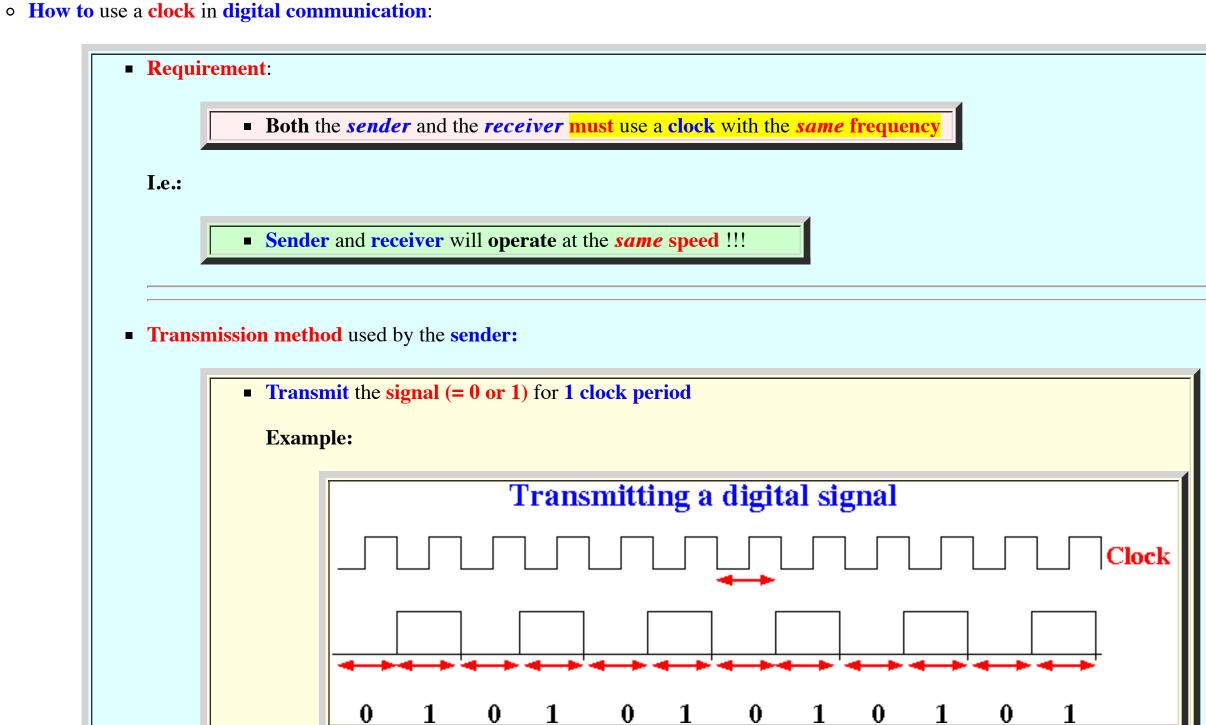


- Answer to the \$64,000 question:
  - You need to know the transmission rate (of the digital signal) in order to decode (= determine the result) a digital signal !!!!!

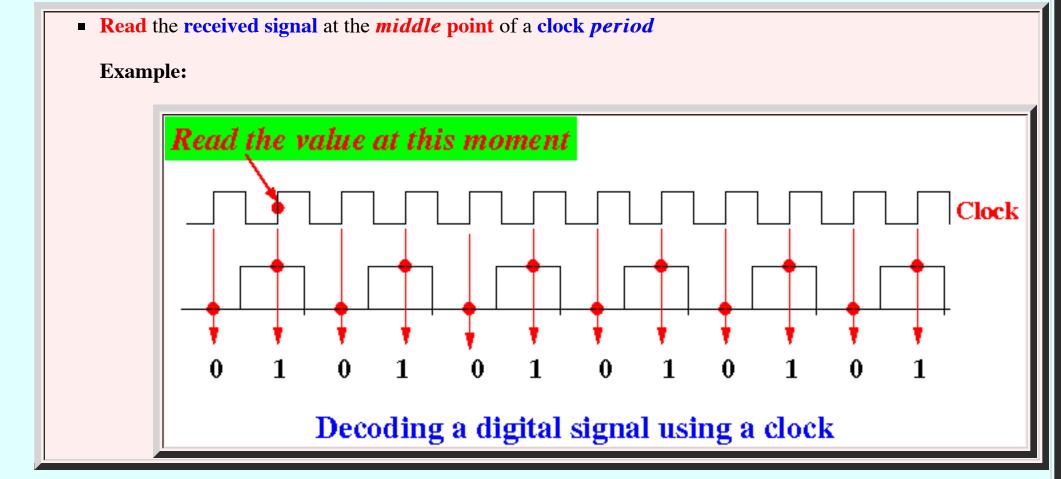
 Clock signal • Fact: ■ Transmission rate (= speed) in electronic devices are controlled by a *clock* signal • Clock signal:



- Clock signal generation circuits:
  - Clock generator = a circuit that produces a timing signal for use in driving (synchronizing) a circuit's operation. See: Wikiopedia
- Transmitting and decoding digital signals



• **Decoding (= reception) method** used by the **receiver:** 

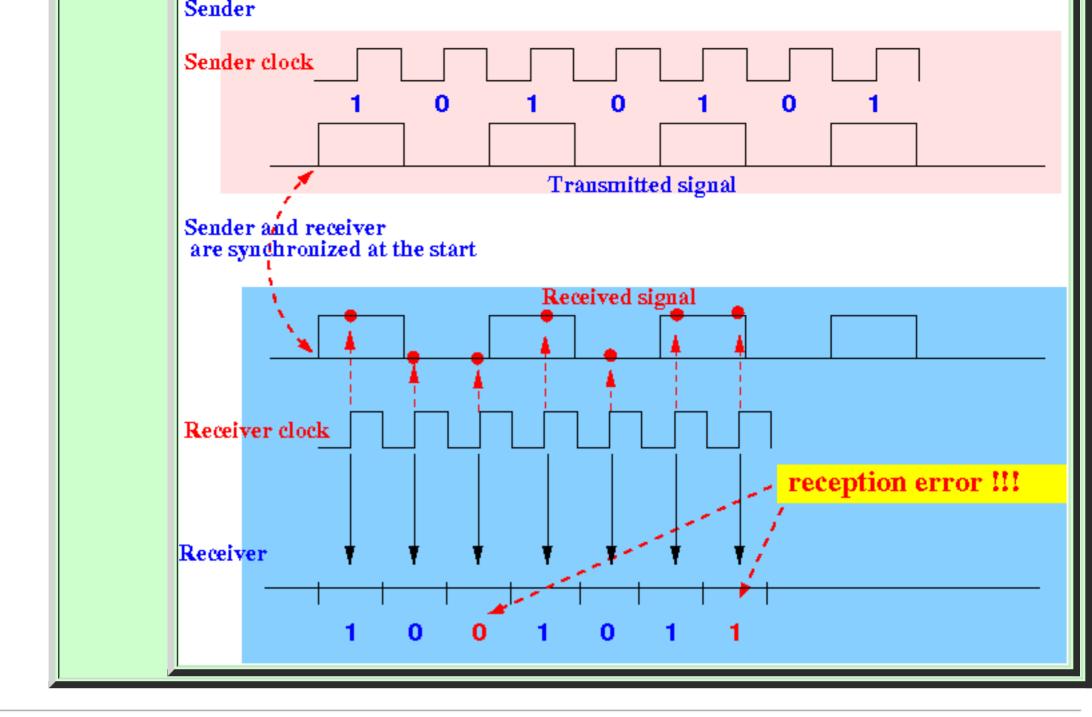


#### • Very important:

■ Sender and receiver that uses digital signaling to communicate with each other *must* use the *same* clock frequency !!!

(Otherwise, you will have *reception* errors!)

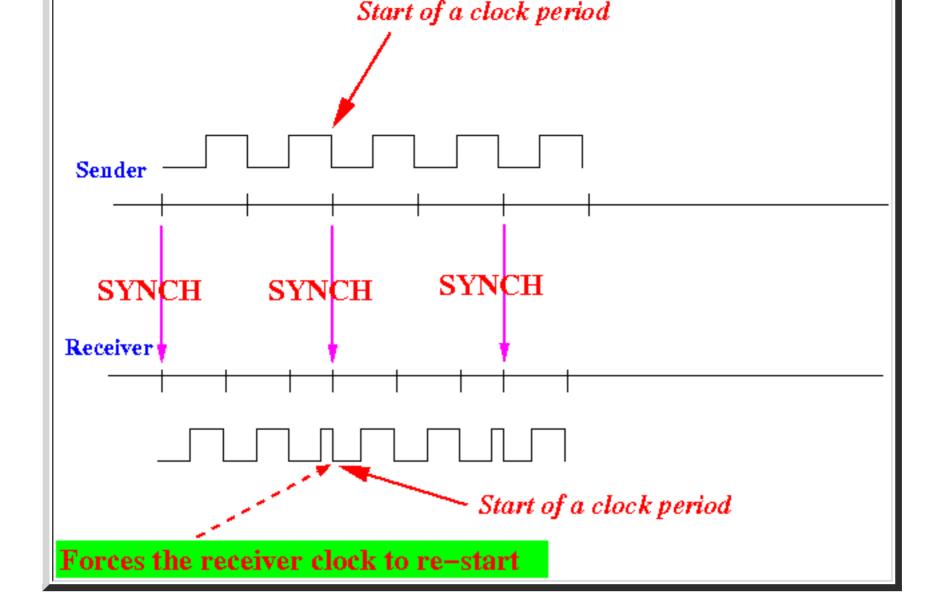
# The problem of clock drift • Click drift • Fact: • No 2 digital clocks will ever run on the same frequency ■ E.g., temperature can affect the frequency (= speed) of digital clocks !!! • Fact of (normal) life: ■ Two "identical" clock signals (used in computer communication) will run at slightly different speeds • The **difference** in **clock speed** is called: "drift". • Consequence of clock drift • Consequence of clock drift: • Clock drift can cause reception (decoding) errors if: A large number of bits is being decoded **Example:**



- Receiver clock (re-)synchronization
  - Important conclussion:
    - The receiver must (from time to time) re-synchronize it clock with the sender's clock

#### **Example:**





• When can you re-synchronize the receiver's clock ???

#### **Problem:**

The receiver can not "see" the sender's clock:

Start of a clock period

Sender

The receiver can only "see" the transmission from the sender:

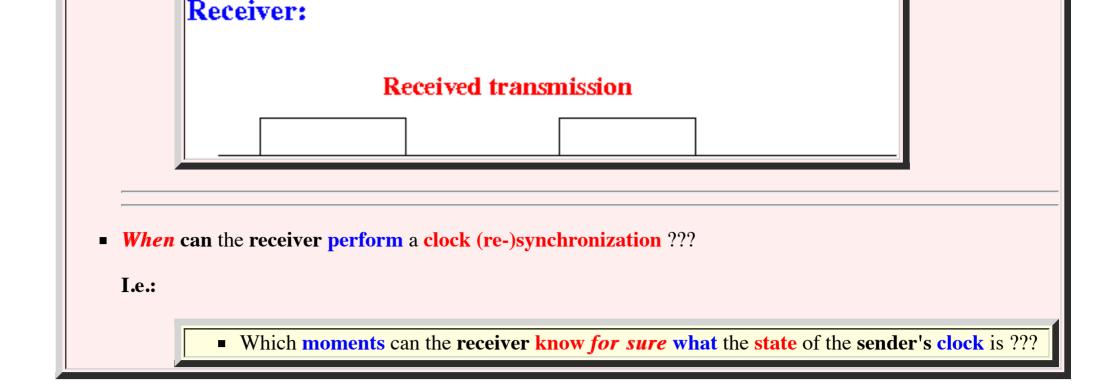
Sender

Clock

1 1 0 0 1 1 Sender's transmission

Receiver "sees" this signal.....

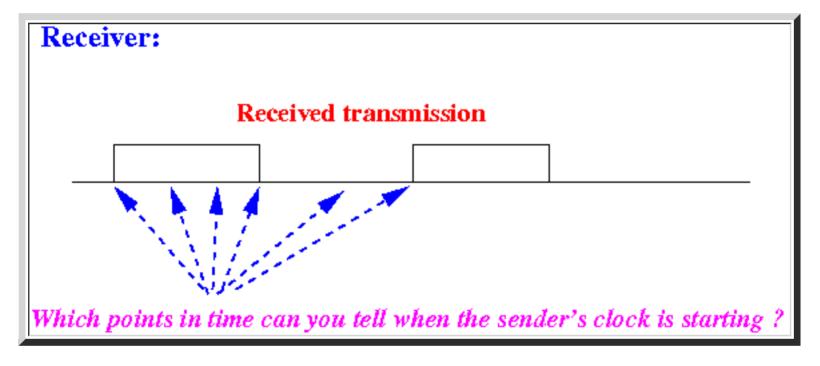
■ This is what the *receiver* will "see":



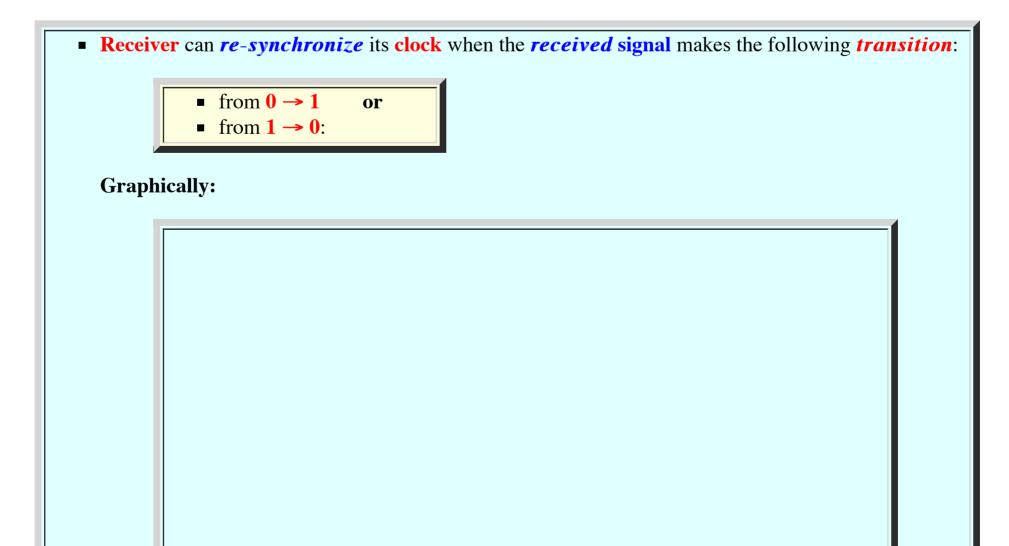
#### • \$64,000 question:

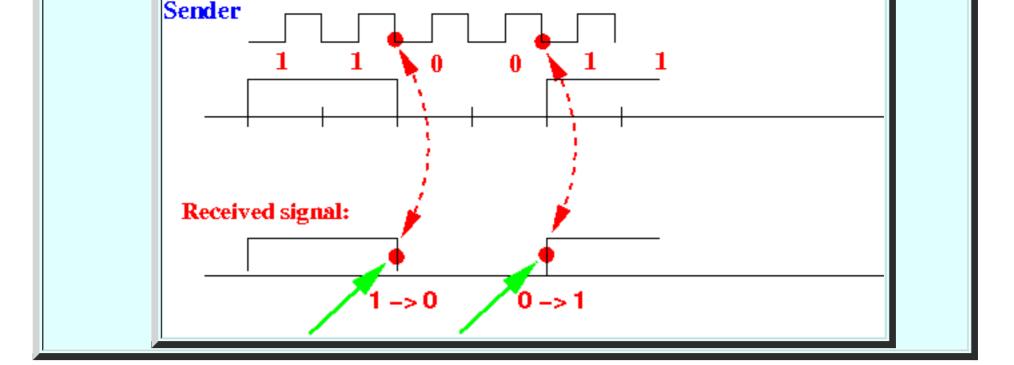
• Which moment(s) in time can you know for sure that the sender's clock is at the beginning of a clock period ???

#### **Pictorially:**

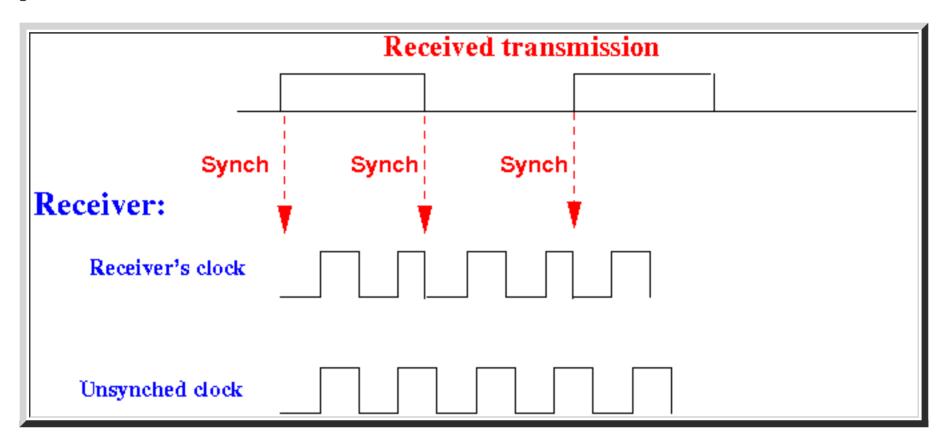


#### **Answer:**





#### **Example:**

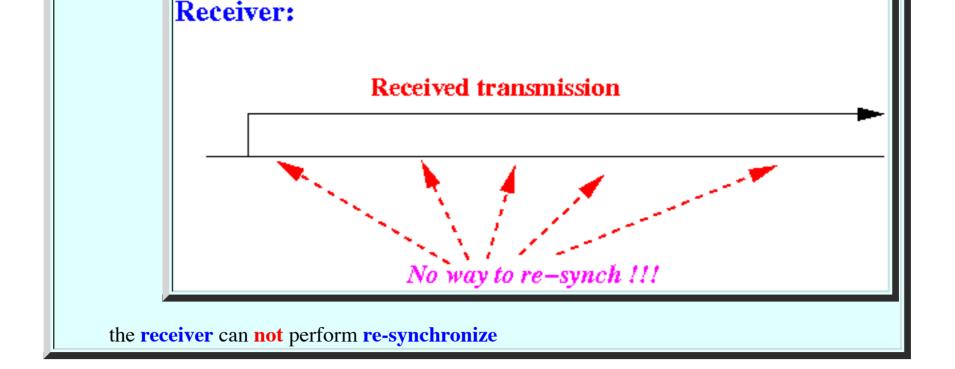


• General practice:

- Each time when the received signal makes a transmission:

   The receiver clock is re-synchronized!!!

  (Re-synchronize means: force the clock signal to restart at the start of a period)
- Wait !!! There is still a problem...
  - There is *still* a **problem**:
    - If the sender transmits a *long* serie of 0's or 1's:



- Solution ???
  - More *complicated* transmission encoding !!!

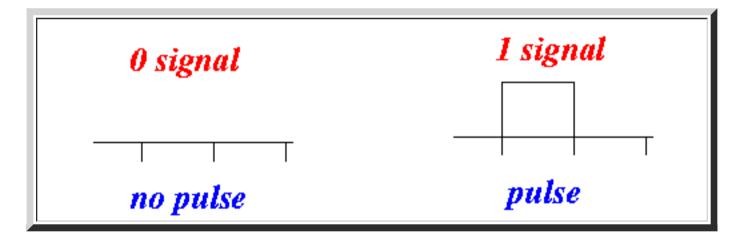
#### The NRZ (Non-Return to Zero) transmission code

- Digital transmission codes
  - There are 2 signaling levels in binary digital transmission:

```
■ low (= 0)
■ high (= 1)
```

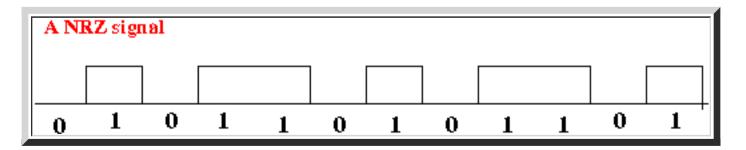
- However:
  - There are *different* ways to *signal* data using these 2 signal levels!!
- We will look at some **commonly used** schemes
  - Each signaling scheme has its strengths and *also*, it's weakness(es)

    (The law of "conservation of misery" --- Herman Bavinck)
- The Non-return-to-zero (NRZ) code
  - The Non-return-to-zero (NRZ) encoding scheme:

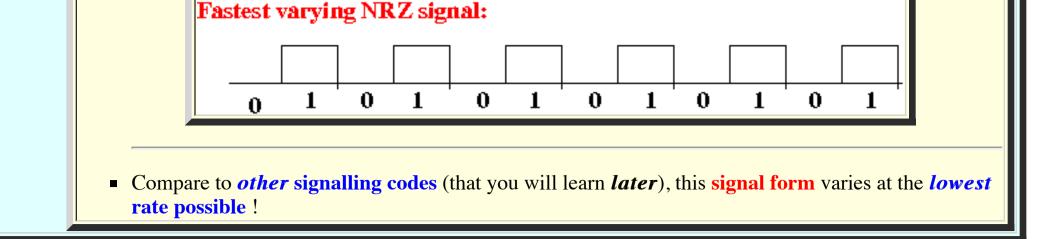


(It's the **most** intuitive code :))

• Example:



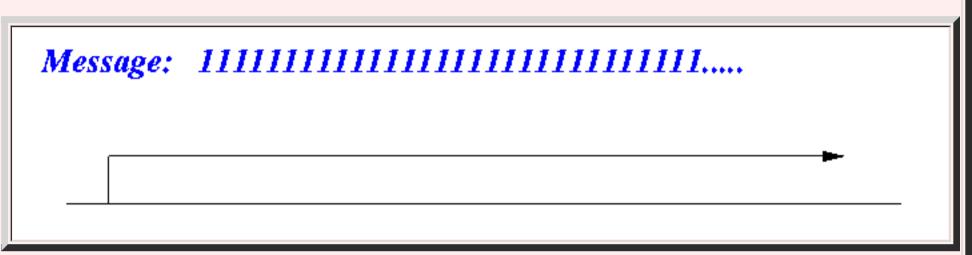
- Advantage:
  - The *highest* frequency of the transmitted signal is relatively *low* Explanation:
    - The **signal** will **vary** at its **fastest** rate when you transmit as string of **0101010101...**:



#### • Disadvantage:

• There are situations where the receiver *cannot* synchronize its clock with the sender's clock:

**Example:** 



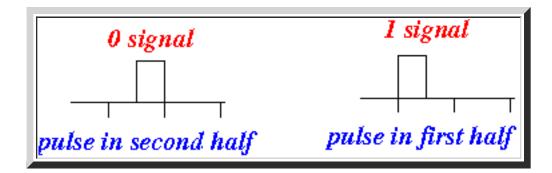
(There is **no transmission** in the signal so the **receiver** can **not perform re-synchronization**!)

#### • Common Practice:

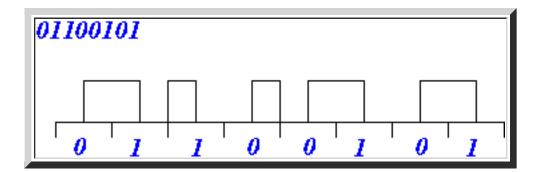
Because of its shortcoming, the NRZ code is only used for very short transmissions (e.g., transmit 1 byte (= 8 bits))

## The *Manchester* transmission code

- Manchester encoding: embedding the clock in the signal
  - The Manchester transmission code:



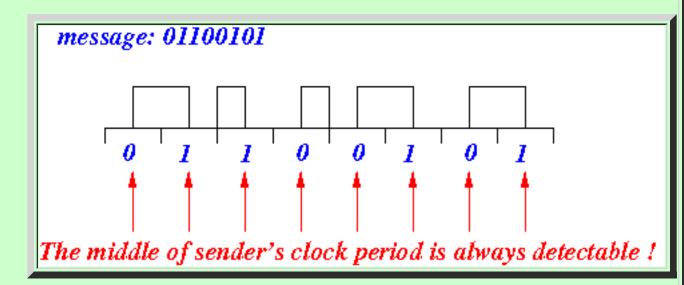
• Example:



• Advantage:

■ The receiver can *always* synchronize its clock at *every* transmitted bit:

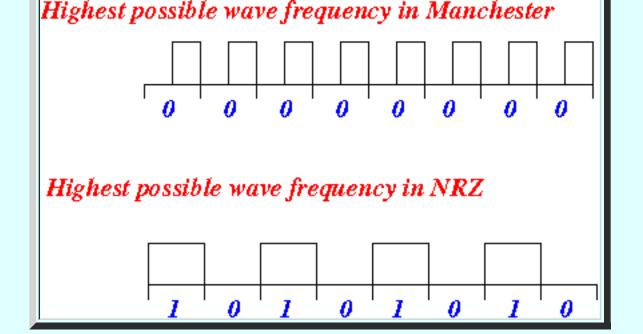
**Reason:** there is a *transition* at the middle of the signal (which is *detectable*)



Therefore:

■ The Manchester code is *ideal* for *very long* digital

transmissions (packet length of several Kbytes).				
<ul> <li>Practice in real life:</li> </ul>				
■ Ethernet packets can be as long as 1500 bytes				
■ Ethernet uses the Manchester code !!!				
• NOTE:				
■ The synchronization operation using Manchester code will reset the receiver's clock to the middle of a clock period				
(It's easy to do, take <b>CS355</b> if you want to learn more about this).				
ne ability to synchronize clocks does not come cheaply				
ntage:				
■ The <i>highest</i> frequency of the transmitted signal is relatively <i>high</i>				
In fact:				
The highest frequency used by the Manchester scheme is twice as large as the highest frequency used in the NRZ method				
■ Example: the highest frequency signal using Manchester code is achieved when you transmit a string of 00000000:				



I drew the **highest frequency** of a **signal** using **NRZ** below in the **above figure**.

You can see the **Manchester code** signal has **2 times** the **frequency** as the **NRZ code** signal.

#### The 4B/5B trnasmission code

- 4B/5B code: fixing the synchronization in NRZ problem cheaply...
  - The **4B/5B code**:
    - 4B/5B code = a very clever way to provide synchronization *opportunities* to the receiver by:
      - Encoding 4 bits using 5 bits
      - The encoding of 5 bits makes sure that there is a transition within the 5 bits
  - **Operation** using **4B/5B code**:
    - The data is *first* transformed using the 4B/5B encoding scheme
    - The **encoded result** is **transmitted** using the **NRZ code** as its **basic transmission** scheme

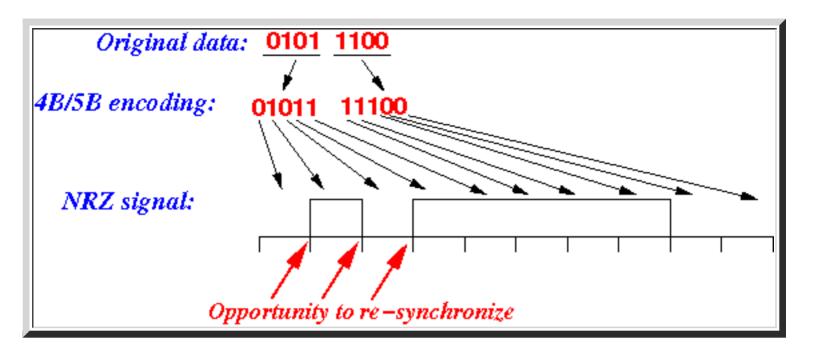
#### Note:

- When the receiver received the transmitted data, it must use the reverse mapping of the 4B/5B encoding scheme to obtain the transmitted data.
- 4B/5B mapping scheme:

Original 4 bits	Transformed 4 bits
0000	11110
0001	01001
0010	10100
0011	10101
0100	01010
0101	01011
0110	01110
0111	01111
1000	10010
1001	10011
1010	10110
1011	10111
1100	11010
1101	11011

1110 11100 1111 11101

#### • Example using 4B5B:



#### Notice that:

- After *transforming* the original sequence of bits, the **result sequence** will have:
  - at most 3 consecutive 0 bits

One way to transmit **3 consecutive 0 bits** is: **0010** (which is transformed to **10100**) followed by **0100** (which is transformed to **01010**)

Original 4 bits	Transformed 4 bits
0000	11110
0001	01001
0010	10100
0011	10101
0100	01010
0101	01011
0110	01110
0111	01111
1000	10010
1001	10011
1010	10110
1011	10111
1100	11010
1101	11011
1110	11100
1111	11101

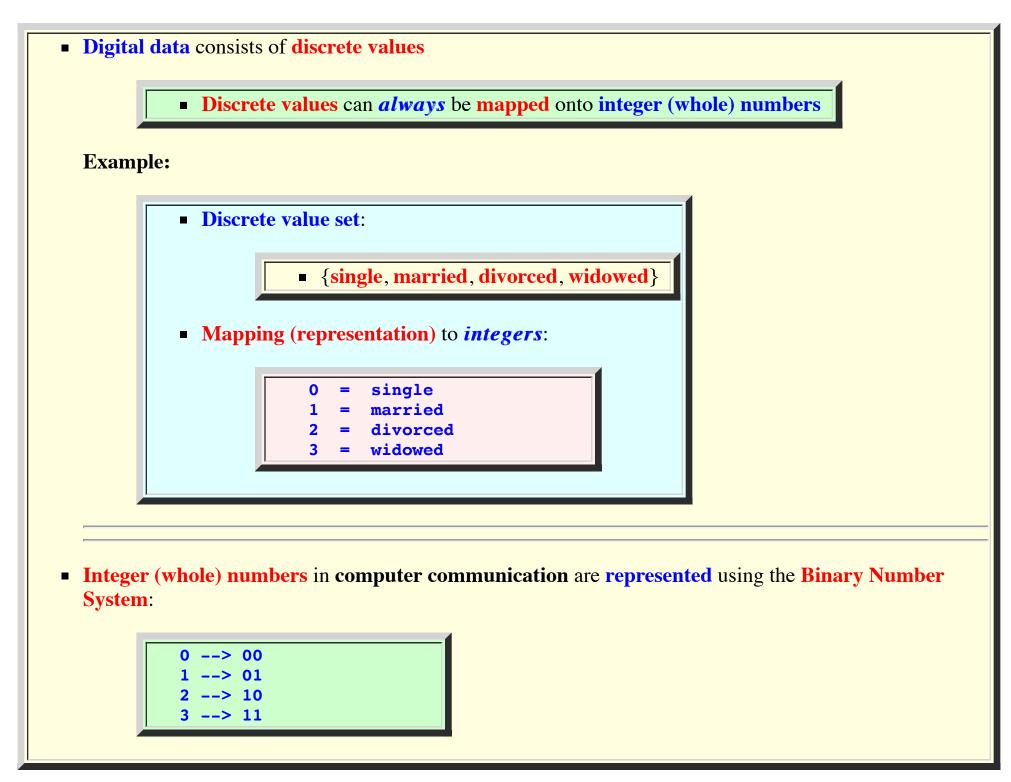
**at most 8** consecutive **1 bits** 

This will happen when you transmit: **0111** (which is transformed to **01111**) followed by **0000** (which is transformed to **11110**)

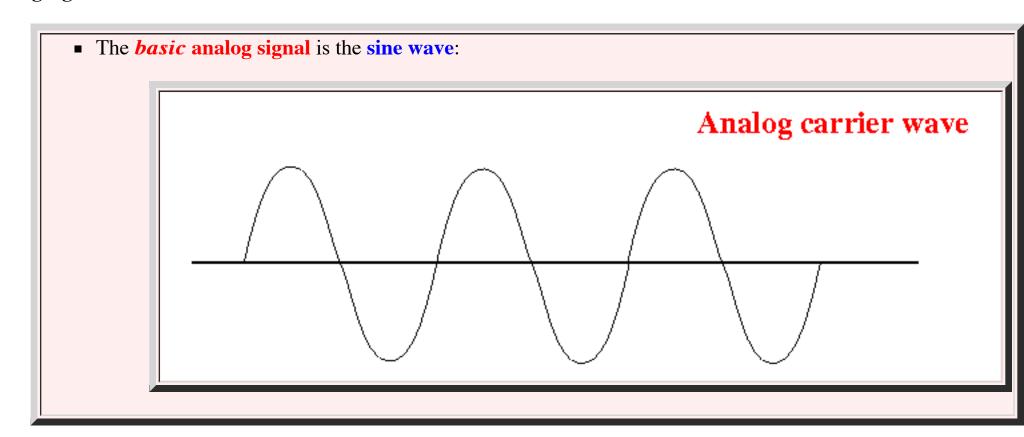
riginal 4 bits	Transformed 4 bits
0000	11110
0001	01001
0010	10100
0011	10101
0100	01010
0101	01011
0110	01110
0111	01111
1000	10010
1001	10011
1010	10110
1011	10111
1100	11010
1101	11011
1110	11100
1111	11101

## Intro: transmitting digital data using analog signals

- Digital data and analog signals
  - Digital data:



• Analog signals:



	■ Modulate (change) the sine wave to represent the values 0 and 1
Recall:	
	<ul> <li>There are 3 modulation methods:</li> <li>Amplitute Modulation (AM): Change the amplitude of the (sine) wave</li> <li>Frequency Modulation (FM): Change the frequency of the (sine) wave</li> <li>Phase Modulation: Change the phase of the (sine) wave</li> </ul>

• Transmitting digital data using analog signals

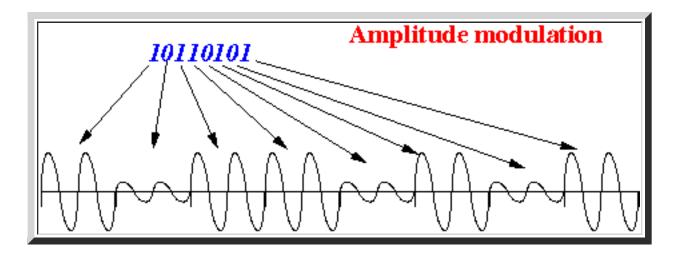
• Transmitting any kind data using analog signal means:

■ Modulate (change) the sine wave using the *input* data

### Transmitting digital data using amplitude modulation

- Transmitting digital data using Amplitude Modulation
  - The amplitude of a wave determines its loudness...
  - Amplitude modulation:
    - **0** = transmit a **softer** signal for one time unit and
    - 1 = a *loud* signal for one time unit

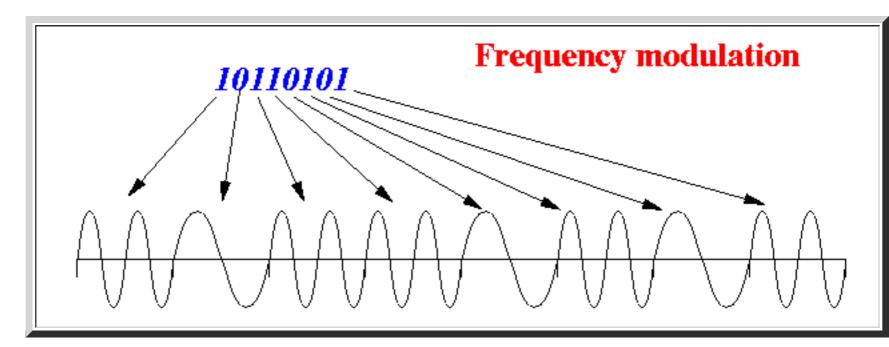
**Example:** (in example, 1 time unit = 2 sine waves)



#### Transmitting digital data using frequency modulation

- Transmitting digital data using Frequency Modulation
  - The **frequency** of a **wave** determines the **pitch** (of the tone)
  - Frequency modulation:
    - 0 = a *lower pitch* tone for one time unit and
    - 1 = a *high pitch* tone for one time unit

**Example:** (in example, 1 time unit = 2 sine waves)

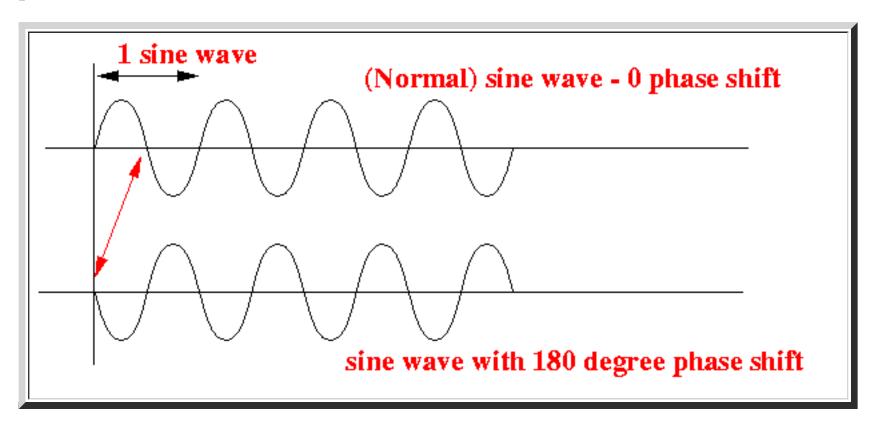


• Cool demo on YouTube on Frequency Modulation: click here

## Transmitting digital data using phase modulation

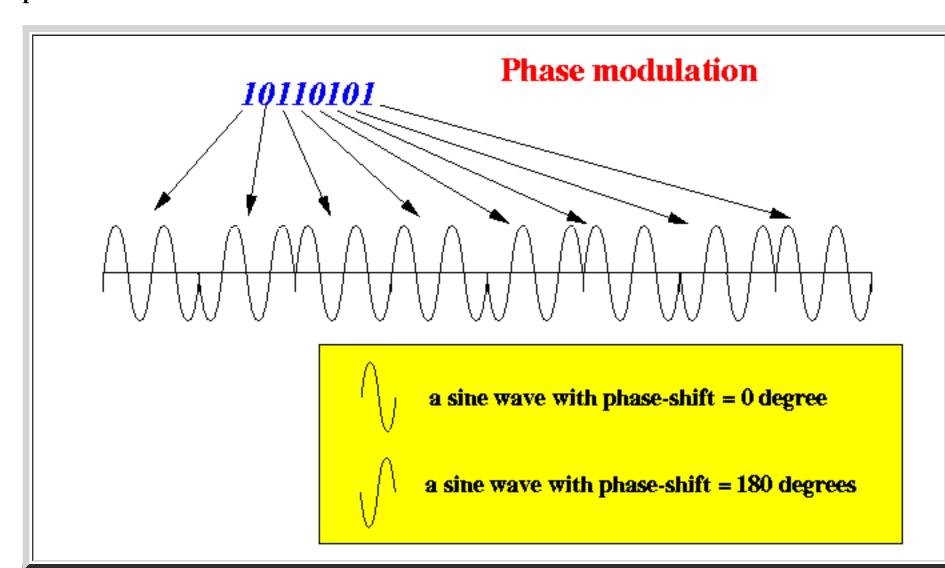
- Transmitting digital data using Phase Modulation
  - The phase of a (sine) wave is the shift in the x-axis direction

#### **Example:**



- Phase modulation:
  - 0 = a sine wave with 0 phase shift for one time unit
  - 1 = a sine wave with 180 degree phase shift for one time unit

#### **Example:**





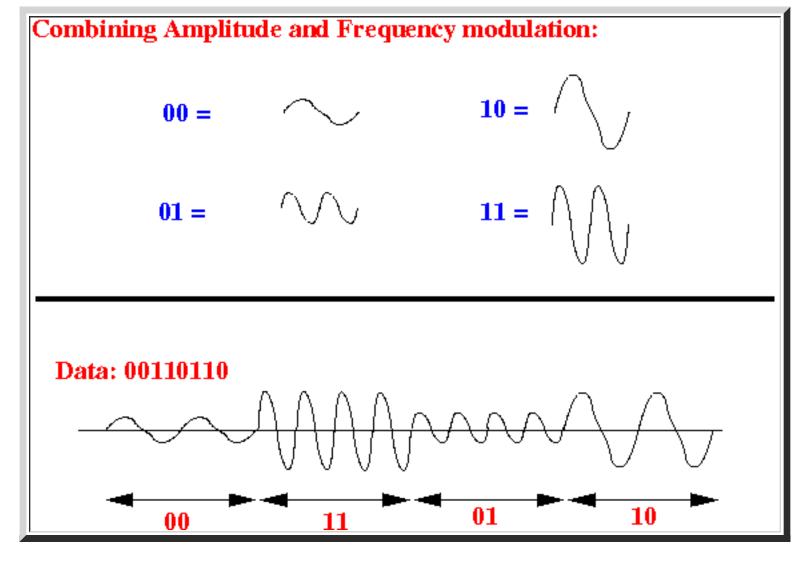
#### Transmitting digital data using a combination of modulation techniques

- Combining modulating techniques in digital data transmission
  - Fact:
- It is common to combine the above 3 modulation methods to achieve a higher data rate

Example: combining amplitude and frequency

00	low amplitude	and	low frequency
01	low amplitude	and	high frequency
10	high amplitude	and	low frequency
11	high amplitude	and	high frequency

#### **Graphical example:**



**Notice that:** 

• We can **now** transmit:

■ 2 bits using 1 sine wave

#### • Historical fact:

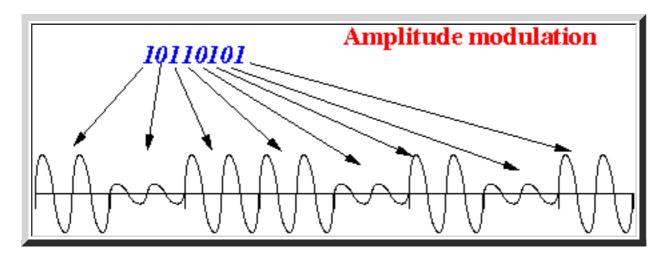
■ *Modems* (which was used around 1990) use a combination of all 3 techniques and can achieve upto 56Kbps:



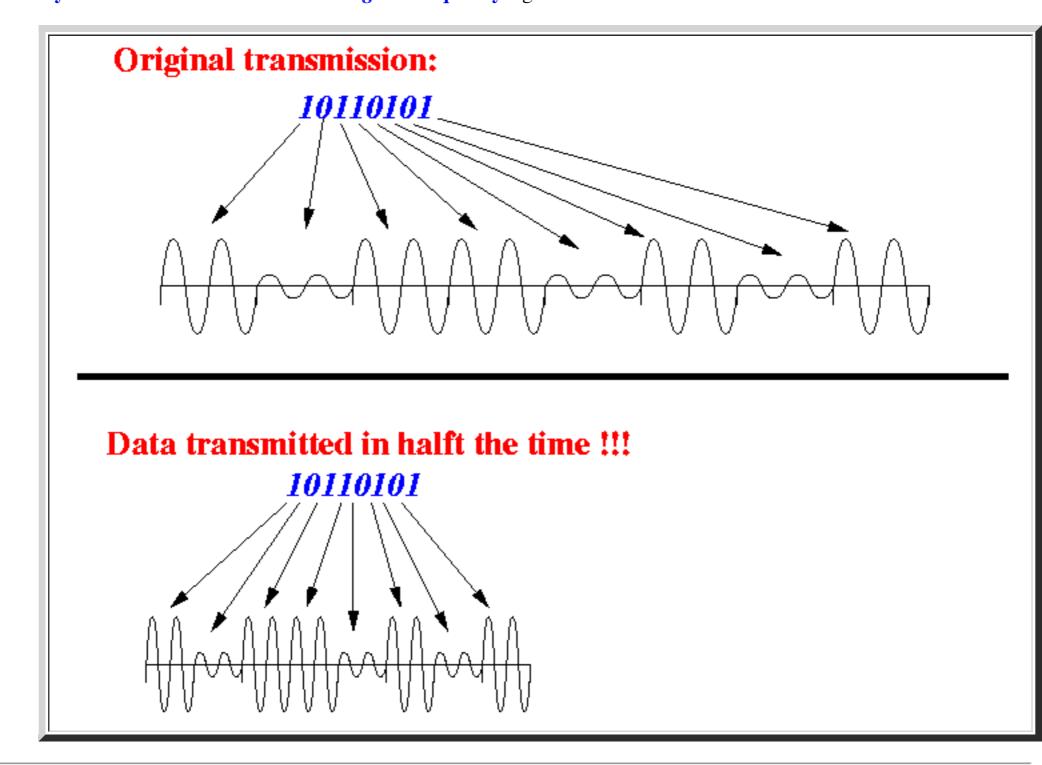
- which was the **maximum** possible data rate on a telephone line

## Bandwidth and maximum transmission rate

- How to transmit data faster
  - **Recall**: transmitting **digital data** using *amplitude* **modulation**:



• One way to transmit data faster is use a higher frequency signal:



• So:

• We can achieve *infinitely* high data transmission rates ???

Obviously not... by why???

- Interesting question....
   Question:
   Why can't we transmit tera bytes of data per second over for example a copper wire ???
  - More *precisely*:
    - What prevents us from transmitting tera bytes of data per second over a copper wire ???

#### Speed limitation in wireless transmissions

- Signal used in wireless transmissions
  - **Wireless** transmission:

Always use electro-magnetic waves
 Note: *light* is *also* a form of electro-magnetic wave !!!

**Example: radio wave** 

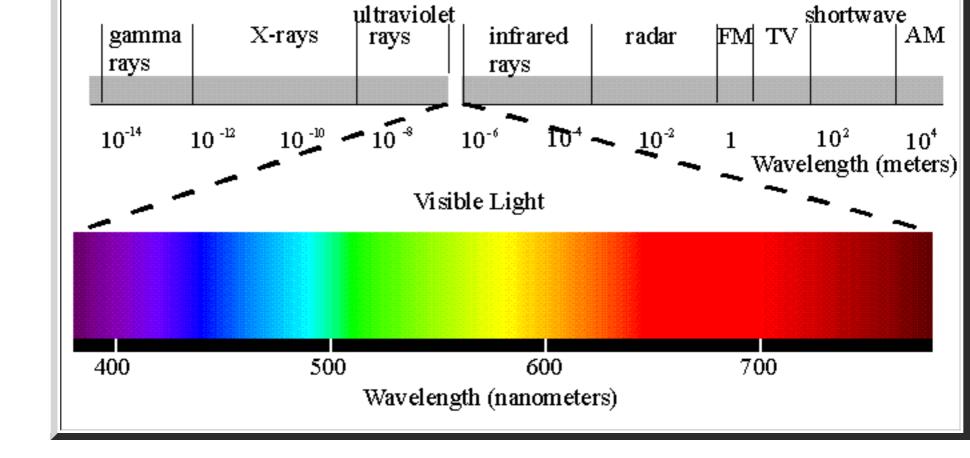


- The *spectrum* of Electro-Magnetic waves
  - Electro-magentic waves goes by many different names, such as:
    - **Radio wave** (this is what **ordinary people** would think of...)
    - Light !!
    - Micro-wave (the kind of electro-magnetic waves used to cook)
    - X-ray
    - And so on....

The *difference* between the electro-magnetic waves above is:

■ Their *frequency*....

• The **frequencies** of **electro-magnetic waves** spans a **very wide range**:



- Transmission data at higher (data) rate
  - Fact:
- Higher data rate transmissions must use a high frequency electro-magnetic wave signal
- Consequence:
  - To achieve *very high* data rate, we would have to use *X-ray* or *gamma-ray*!!!
    - These rays are *harmful* to biological entities!!!!

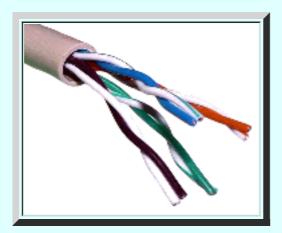
- Breaking news....
  - Cell phone and brain cancer:
    - Be careful with your **cell phones**:
      - Cellphone radiation can cause cancer: <u>click here</u>

(Cellphone operates at frequencies that is *very* close to *micro-wave* --- the kind of waves used to *cook* food....

#### **Transmission media**

- Commonly used Transmission media
  - Commonly used wired transmission media:

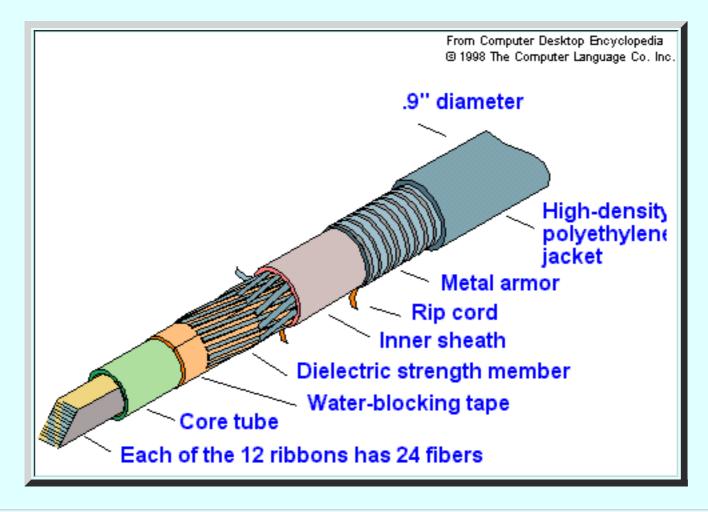




■ Coaxial cable - (better)



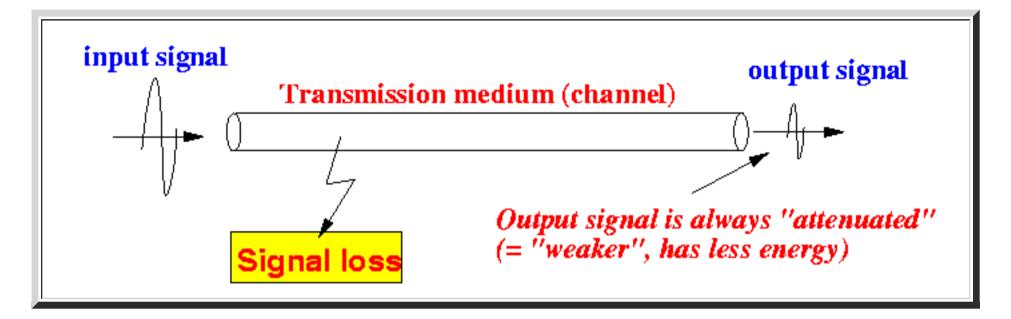
Optical Fiber (best)



#### Transmission over a copper wire

- Energy loss in signal transmission in copper wires
  - Physics:
- When an electrical signal (= electricity) traverse through a copper wire, the always lose some power

#### **Graphically:**



#### **Terminology:**

- Attenuation = loss of power
- Source of power loss
  - There are 2 ways that an electrical signal can lose power:
    - **Resistance** (in the wire):



**Resistance** in the wire will cause **some** of the **energy** in the **signal** to be **converted** into **heat** 

Radiation:



When the **frequency** of the **electric current** is **very high**, an **alternating current** will generate a **strong electro-magnetic** field.

The wire will work like an antenna - and some energy of the signal will radiate away....

- Fact from Physics:
  - The **amount** of **power loss** depends on:
    - The *resistance* of the transmission medium
    - The *frequency* of the electro-magnetic signal (= how much of the power will radiate away).
- Energy loss characteristics
  - The energy loss characteristics is as follows:
    - When the (*alternating*) electric current has a very *low* frequency:
      - the current encounters high electrical resistence.
         (DC current generates a low of heat !!!!)

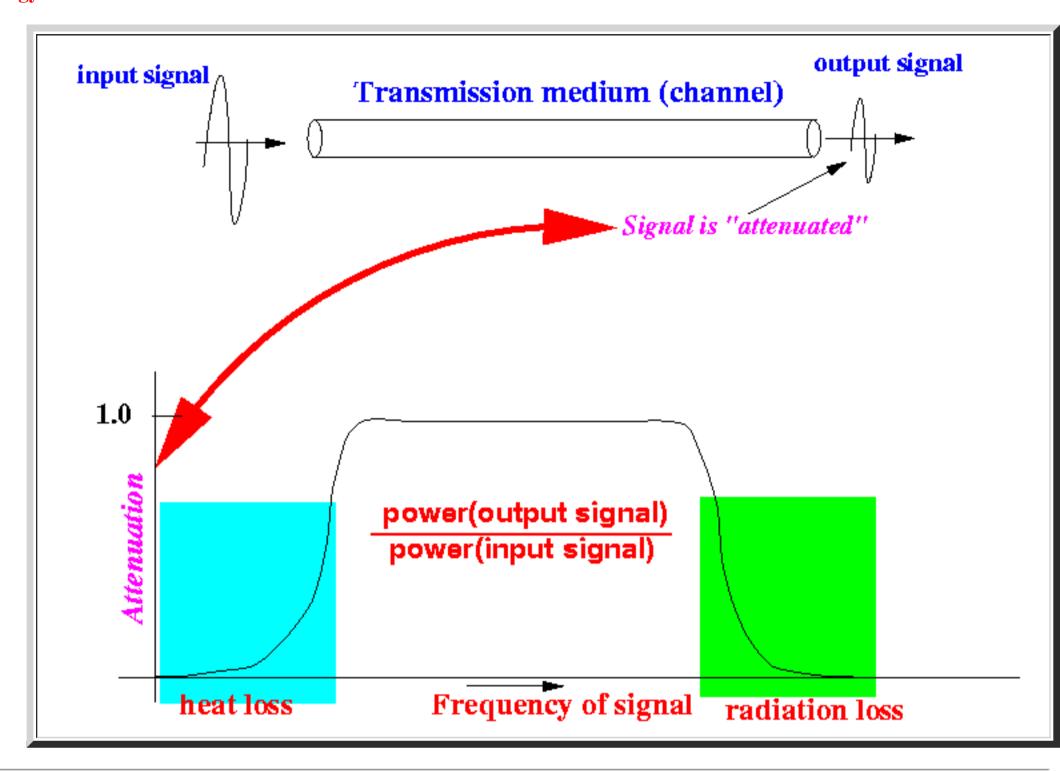
#### **Result:**

- A large portion of the energy in the transmitted signal is lost because the energy is converted into *heat*
- When the (*alternating*) electric current has a very *high* frequency:
  - the current will generate a strong electro-magnetic field.

#### **Result:**

■ A large portion of the energy in the transmitted signal is lost because the energy is converted into electro-magnetic wave (radiation)

• Energy loss characteristic of a wire transmission medium is as follows:



## The bandwidth of a transmission medium

- Acceptable level of power loss
  - Acceptable loss threshold (set by electrocal engineers):

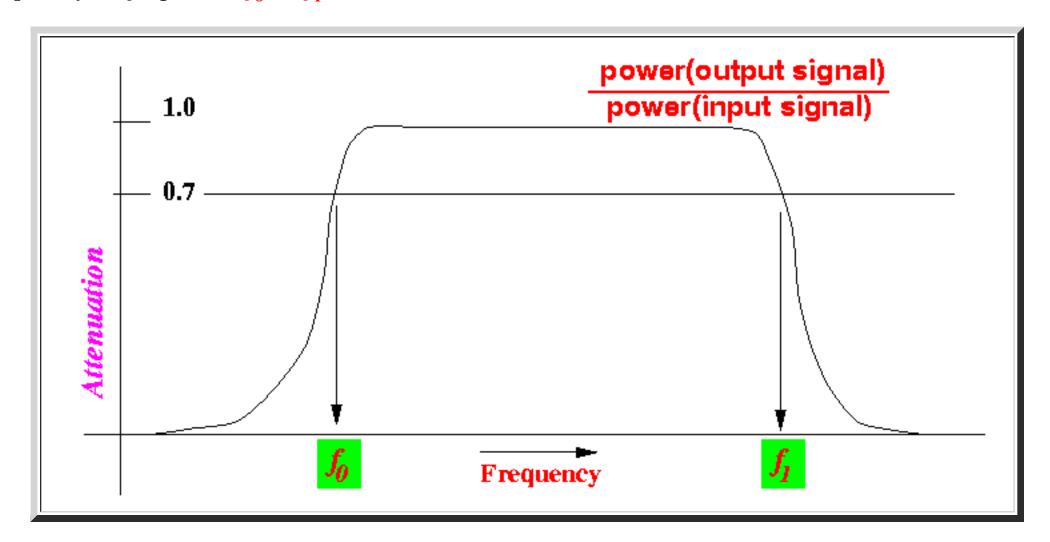
```
power(output signal)
----- ≥ ln 2 ~= 0.7
power(input signal)
```

This **value** is known as the **half-power point**: <u>click here</u>

- Range of *acceptable* frequencies of a transmission media
  - Definitions:

```
    Let f<sub>0</sub> = the low end frequency such that:
    power(output signal)
    Let f<sub>1</sub> = the high end frequency such that:
    power(output signal)
    power(output signal)
    power(input signal)
    power(input signal)
```

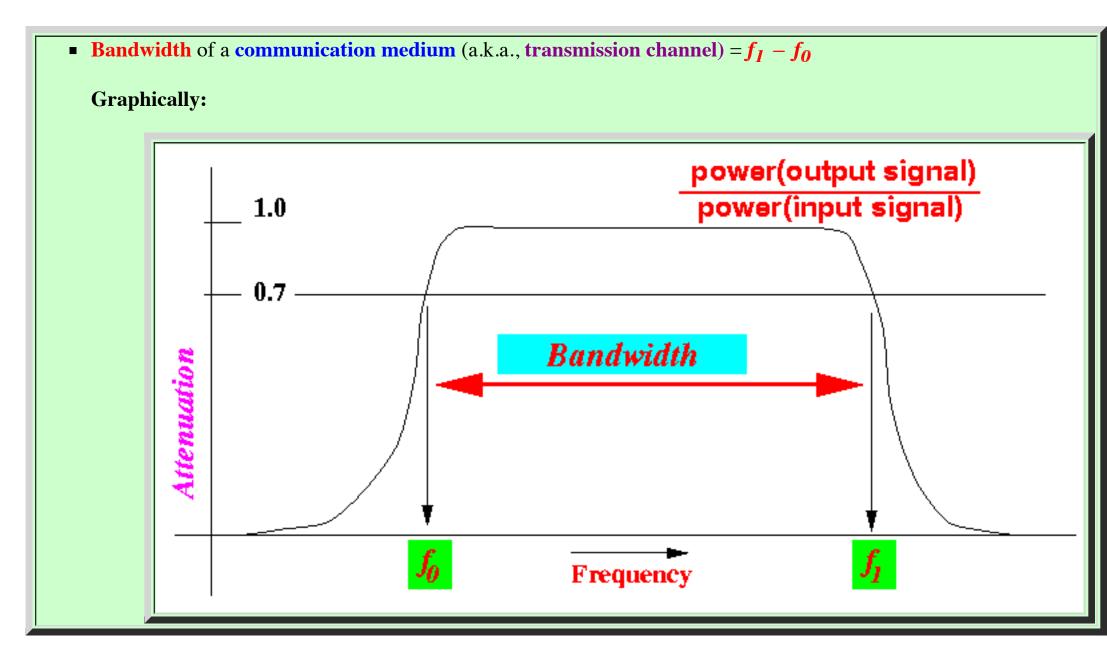
• Graphically: the *frequencies*  $f_0$  and  $f_1$  are defined as follows:



• Fact:

• Electrical signals (= alternating current)) that have a frequency between  $f_0$  and  $f_1$  will pass through the

- Bandwidth of a communication medium
  - **Definition: bandwidth** of a **communication medium**:



**Fact:** 

- There is a direct relationship between
  - Bandwidth of a communication channel

and

Maximum transmission rate of the communication channel (= how fast you can transmit communication channel)

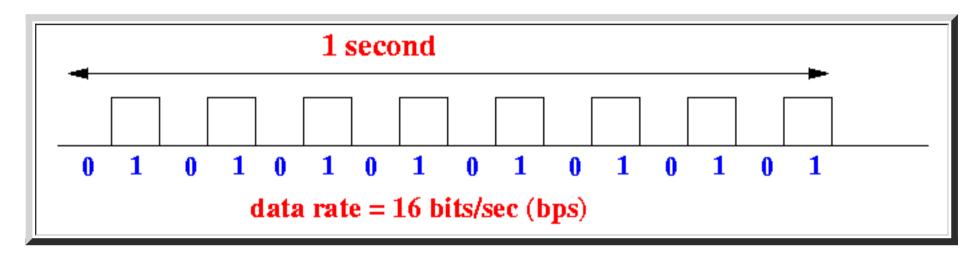
This **relationship** is called the **Shannon formula** and will be discussed **next** 

#### Data transmission rate and frequency

- Data (transmission) rate
  - Fact:
- Computer communication often use *digital* signalling

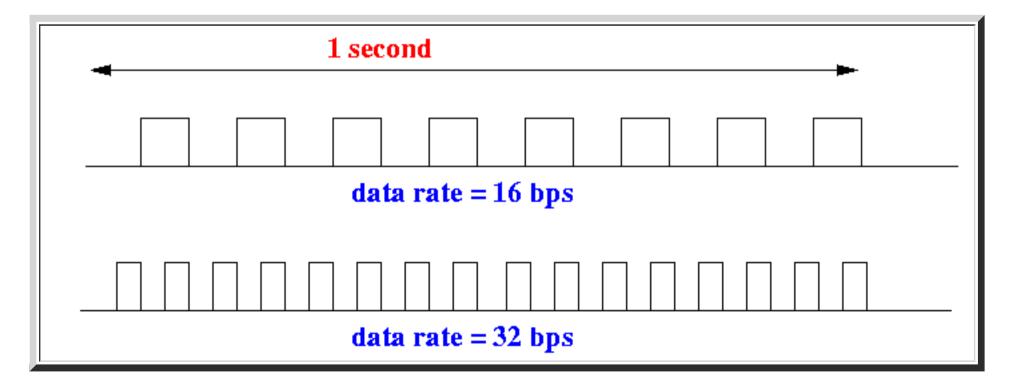
  (I.e., computers transmit a series of pulses that represent bits)
- Definition: data rate
  - data rate (or data transmission rate) = number of bits transmitted in one second
  - Unit is: bits/sec

#### **Example:**



- Data rate and frequency
  - Observable fact:
    - Faster data rate will result in a signal with a higher frequency

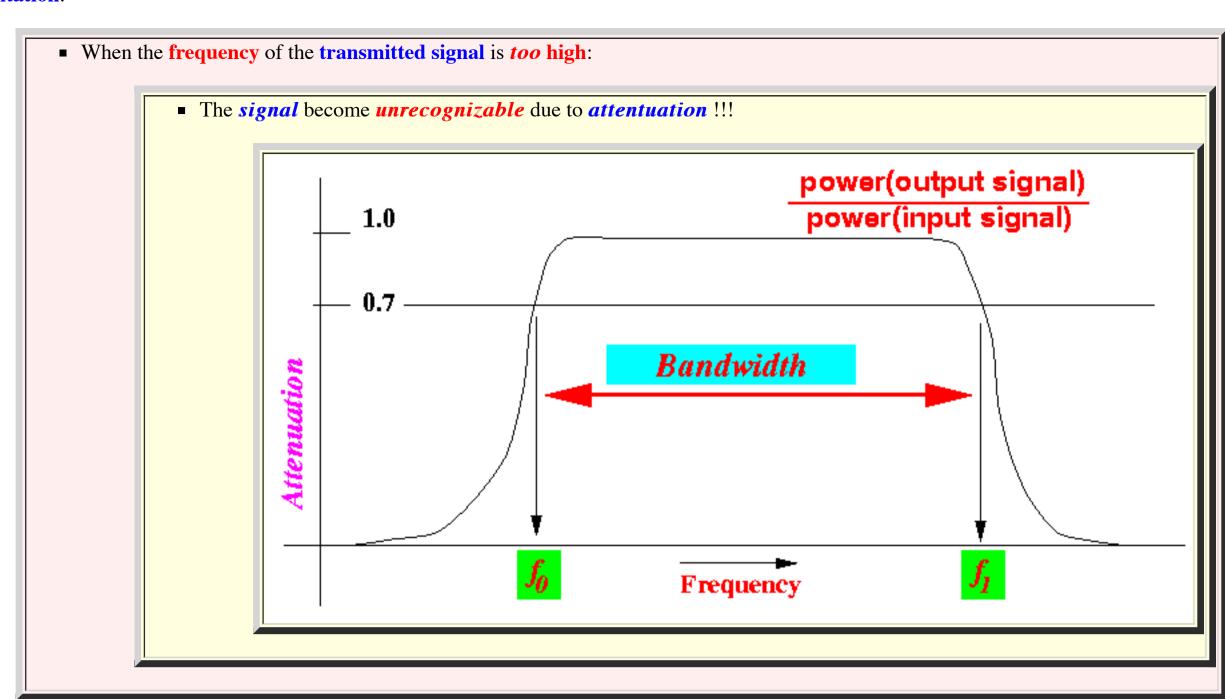
#### **Example:**



The **frequency** of the **second signal** is *higher* (more cycles per second)

- Relationship between *bandwidth* and data transmission *rate* 
  - Clearly:
- A higher data (transmission) rate wil result in a signal with higher frequencies

#### • Limitation:



#### The Maximum data rate of a transmission medium

- Another factor that affect reception quality
  - Fact:
- Beside *physical* constraints imposed by the property of the the transmission media, another factor that affects the quality of the received signal is:



- Maximum data rate of a transmission channel
  - The **factors** that **determine** the **data transmission rate** over a **transmission medium** (from electronics):
    - **Bandwith** of the **transmission medium** (i.e., the **physical property**)
    - Ambient noise
  - Theoretical maximum data rate of a transmission channel: (no proof)
    - **Maximum** data rate of a transmission medium is:

$$C = B \log_2 \left( 1 + \frac{S}{N} \right)$$

where:

```
C = maximum data rate (in # bits/sec) --- a.k.a. capacity

B = Bandwidth of the transmission medium (in Hz)

S = power of the transmited signal (in Watt)

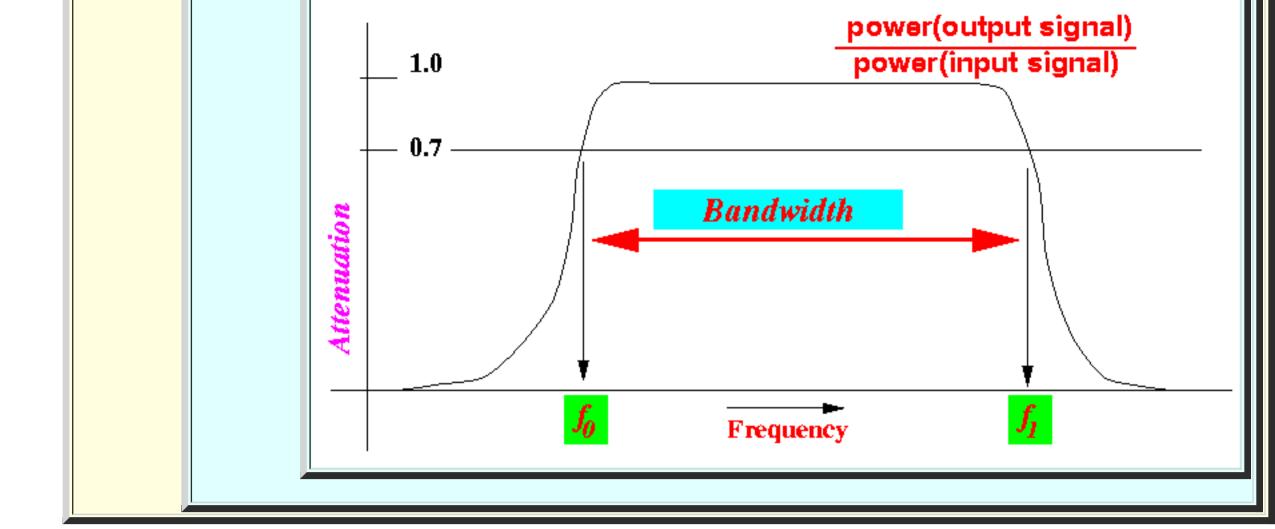
N = power of the ambient noise (in Watt)

S/N is called the "Signal to Noise ratio"
```

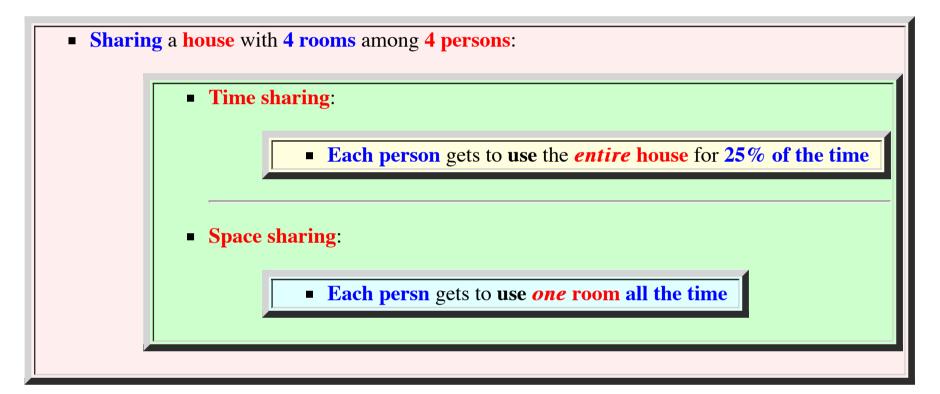
Note:

This famous in click here	result is derived	by the late Math	ematician <mark>Cla</mark> ı	ide Shannon (s

# **Sharing a Transmission Medium** • Motivation to *share* a communication medium • Fact: ■ The capacity (data rate) of a transmission medium is very large **Example:** • One optical fiber wire can transmit ≥ 1 TeraBits/sec!!! (See: click here) ■ A *single* user can not fully use 1 TeraBits/sec data rate... • Common practice: ■ Multiple users usually share one single communication channel Terminology • Multiplexing: multiplexing = the techniques used to share a communication medium • Multiplexing Techniques • There are 2 commonly used multiplexing techniques used to share transmission capacity in the physical layer (= signal transmission layer): ■ The transmission channel can be shared in the *time* dimension: • *Time* division multiplexing ■ The transmission channel can be shared in the "space" (frequencies) dimension: • Frequency division multiplexing **Note:** ■ The "space" is a communication medium is the range of frequencies inside bandwidth of the transmission medium:

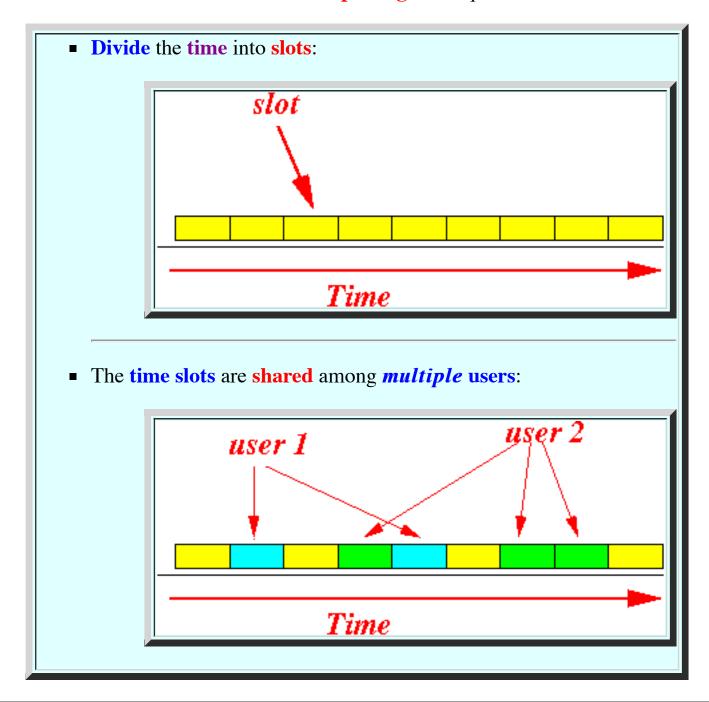


#### • Analogy:

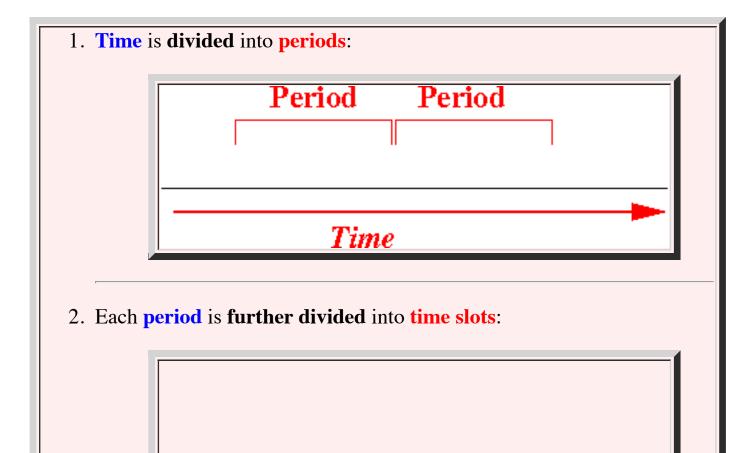


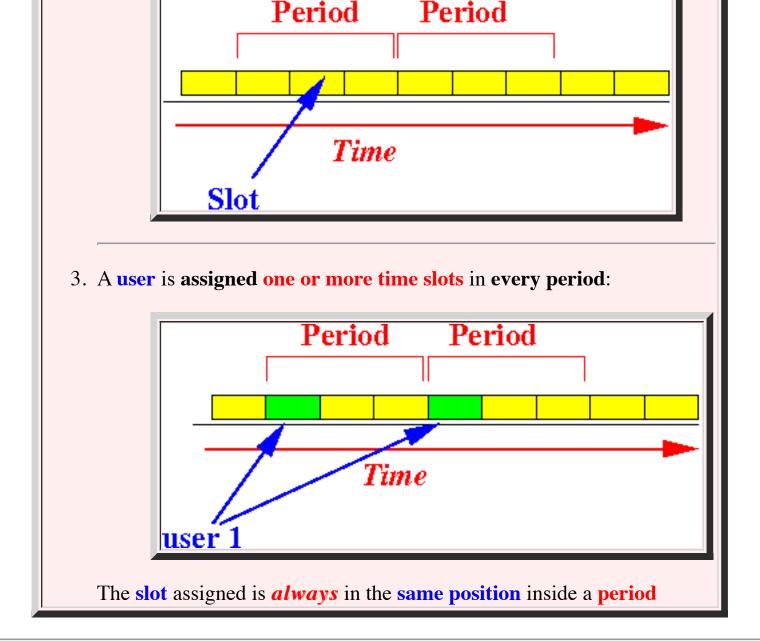
## Synchronous/Asynchronous Time Division Multiplexing

- Sharing a channel in temporal dimension: Time Division Multiplexing (TDM)
  - The basic idea of the *Time* Division multiplexing technique:

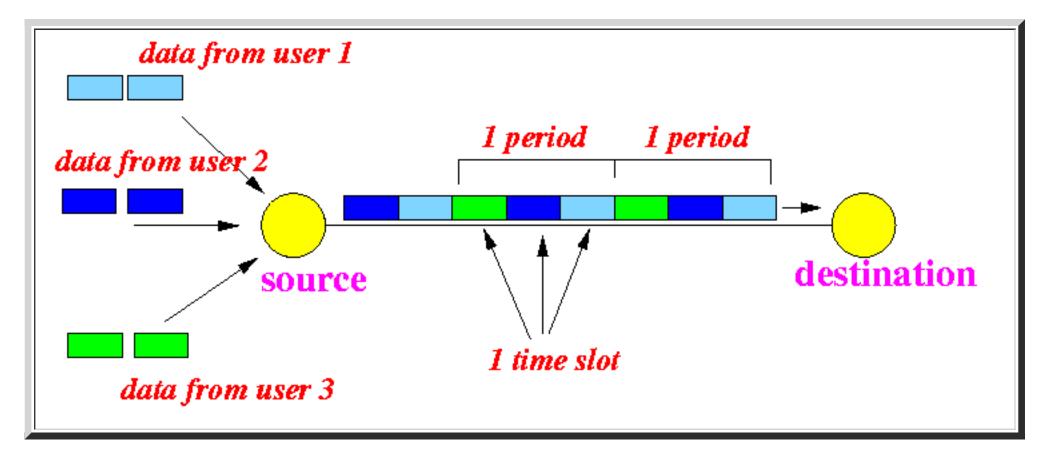


- The **2 variants** of the **TDM** technique:
  - **Synchronous TDM**
  - Asynchronous TDM
- Synchronous TDM
  - The **Synchronous TDM** technique:





#### **Example:**



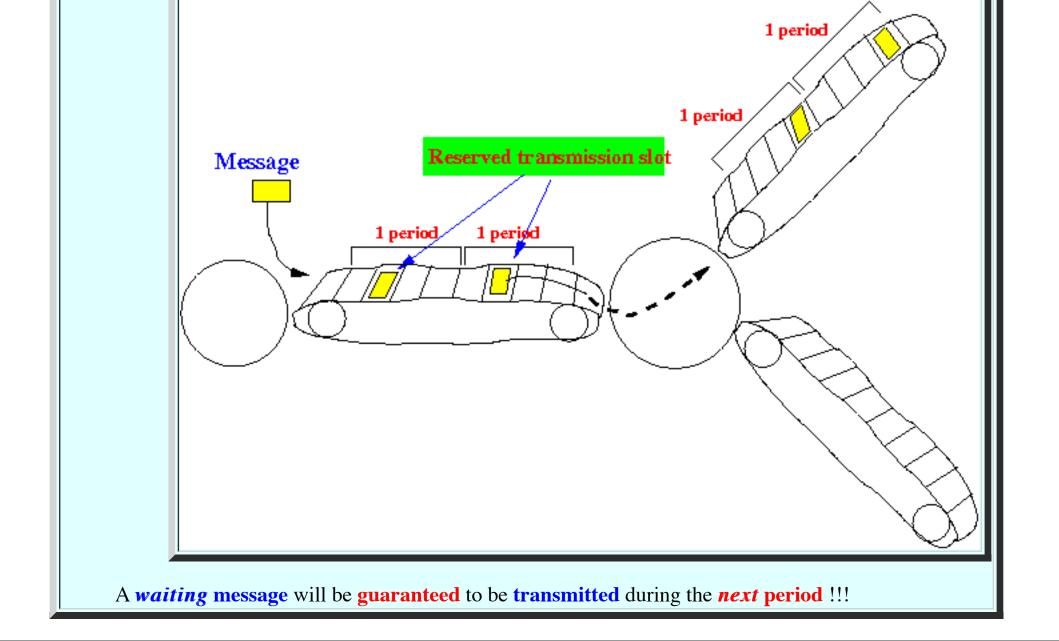
- Advantage/disadvantage of Synchronous TDM
  - Advantage of Synchronous TDM:
    - Simple (= easy) to implement

       Synchronous TDM can provide performance guarantee

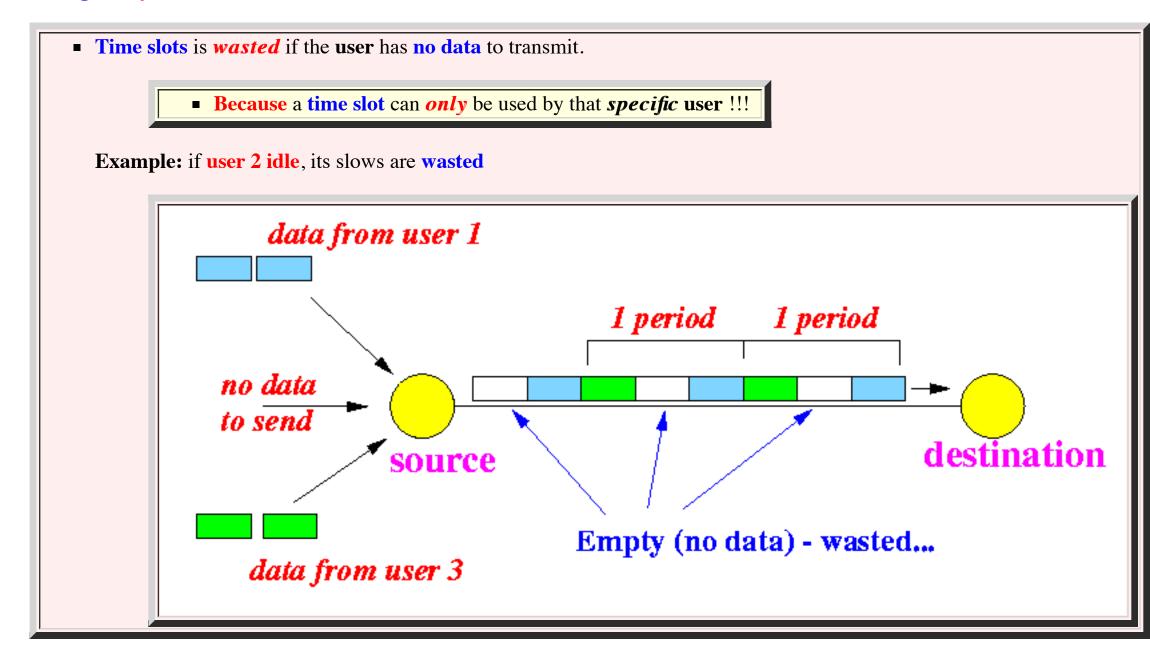
       Asynchronuous TDM can provide a fixed waiting time guarantees

      (The waiting time is at most 1 period !!!)

      Because the Synch. TDM method works like a converyer belt:

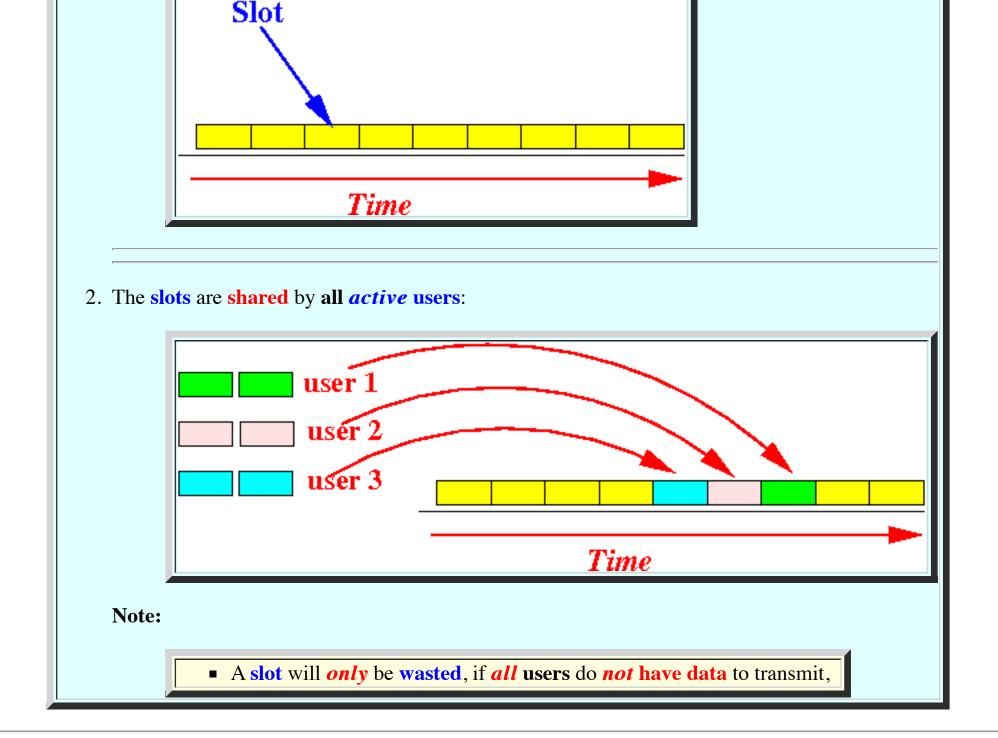


• **Disadvantage** of **Synchronous TDM**:

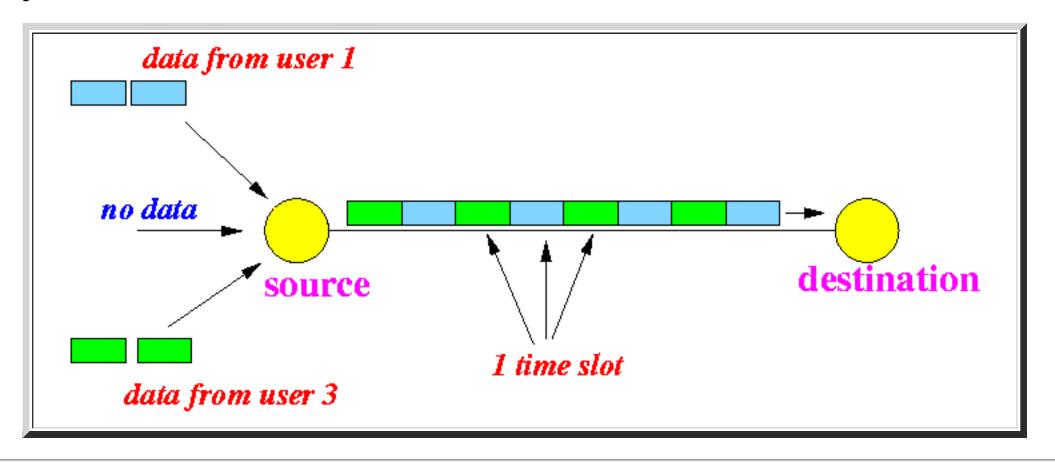


- Asynchronous TDM
  - The **Asynchronous TDM** technique:

Γ	1. Time is divided into slots:



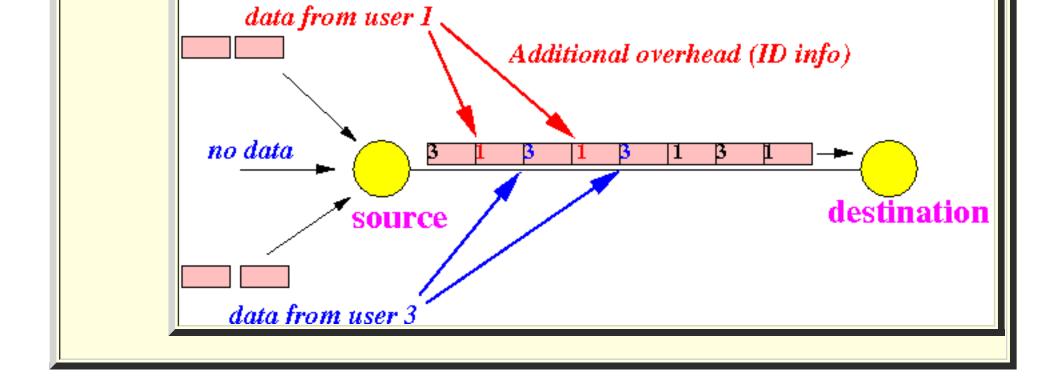
# **Example:**



# • Implementation note:

■ I used colors to indicate the sender of the packets....

■ In reality, the packet are distinguished by a source/destination ID:



- Advantages/disadvantages of Asynchronuous TDM
  - Advantage of Asynchronuous TDM:
    - Efficient use of the transmission capacity.
      - Asynchronuous TDM will *only* waste transmission capacity (= slots) when *all* senders are idle
  - **Disadvantage** of **Asynchronuous TDM**:
    - It's much *harder* to provide service guarantees because:
      - The time that a user gets a transmission slot depends on the *current* number of active users
      - Asynchronuous TDM can not provide a fixed waiting time guarantee

Note:

■ Asynchronous TDM can provide a soft guarantee on delay (waiting time) using flow control

Flow control = controlling the *transmission rate* of data over a communication link

# **Frequency Division Multiplexing**

• Sharing a ch	nannel by "space": Frequency Division Multiplexing
o Frequ	nency Division multiplexing:
	■ The range (space) of frequencies is divided among the users
• Fact fr	<ul> <li>The range (space) of frequencies is divided among the users</li> <li>The signals sent using different frequency ranges do not interfere with each other</li> <li>Radio <ul> <li>Transmissions on FM 98.5 do not interfere with transmissions on FM 94.1</li> </ul> </li> <li>th Frequency Division Multiplexing</li> <li>Users that use different frequencies to transmit can not receive each others' transmissions!!!</li> </ul>
<ul><li>Example</li></ul>	ple:
	■ Transmissions on FM 98.5 do <i>not</i> interfere with transmissions on
• Problem wit	th Frequency Division Multiplexing
<ul><li>Proble</li></ul>	em:
Result	i:
	■ Freq. Div. Multiplexing is <i>not</i> used in computer communication!!!
• Note:	
	■ Frequency division multiplexing is mainly used in radio or walkie-talkie

# Intro: Error Detection and Correction at the Physical Layer

- Errors at the Physcial Layer:
  - Fact:
- A transmitted signal can be received incorrectly
   Example:
   A "1" signal can be received (and decoded) as a "0" signal
- Common causes of transmission errors:
  - Background noise:

     Radiation (from space) can cause single bit errors in a stream of transmitted bits

     Weather related (i.e., lightning):

     Lightning can cause multiple consecutive bit errors in a stream of transmitted bits
- Typical error rates
  - Typical error rates:
    - Error rate in twisted pairs (copper wires):
       1 bit error in every few 1000 bits transmitted.
       Error rate in optical fibers:

o Dete	<ul> <li>ting transmission errors:</li> <li>Error detection is a part of the function to provide reliable communication</li> </ul>
	<ul> <li>The layer (see OSI model: click here) that is responsible for providing reliable communication is:</li> <li>The data link layer         <ul> <li>(And not the physical layer)</li> </ul> </li> </ul>
	■ Technically, it is <i>not</i> necessary to detect errors at the Physical Layer:
• Why	detect errors at the <i>physical</i> layer:
	■ Early error detection in the Physical layer will result in faster response
	■ It is also more efficient:
	■ The sooner we detect an error (and stop the processing of a message), the less computation resources we will waste

# Principle of error detection and correction

- Illustrative question
  - Example of *transmitted* data:

11111111

• Example of *received* data:

11110111

### **Question:**

- How can we tell whether some bit(s) is/are incorrect (without know what the transmitted data was) ???
- Principle of error *detection/correction* methods
  - Principe to detect/correct of errors:
    - 1. Embed extra bits into the transmitted message
      - The *insert* extra bit creates a certain *pattern* in the bits stream (usually conforming to some (mathematical) property)
    - 2. The receiver checks for the embedded pattern in the received data:
      - Pattern was found ⇒ message assumed to be correct
      - Pattern was *not* found  $\Rightarrow$  message assumed to be in *error*

• Mathematics: coding theory

• FYI:	
	<ul> <li>The mathematical theory that deal with design patterns in the transmission is called coding theory.</li> </ul>
I will o	nly discuss some commonly used codes
mmonly u	sed error detection/correction schemes
	only used schemes:
	■ Error detection schemes:
	<ul> <li>Parity checks</li> <li>Cyclic Redundancy Check (CRC)</li> </ul>
	- Error Correction Schemes:
	<ul> <li>Hamming's code (can correct 1 bit error)</li> <li>Solomon-Reed's code (can correct 2 bit errors)</li> </ul>

# **One-dimensional parity-based error detection schemes**

- Parity
  - Parity:
- Parity = the state of being equal (from dictionary)
- Parity-based error detection scheme:
  - Add some bits to the transmitted message so that:
     The number of 1 bits in the message is even (or odd)

(That is an exmaple of a *pattern* that I was eluding to in a previous webpage....)

• Error detection using a parity based scheme:

- When a message is received, the reciever checks for:
   the parity property
   to verify if there was an error.
- One-dimensional Parity Scheme
  - Even and Odd Parity:
    - Even parity:
       Add one extra bit to the message so that the total number of 1's in the message is always even
       Odd parity:
       Add one extra bit to the message so that the total number of 1's in the message is always odd
  - Examples:

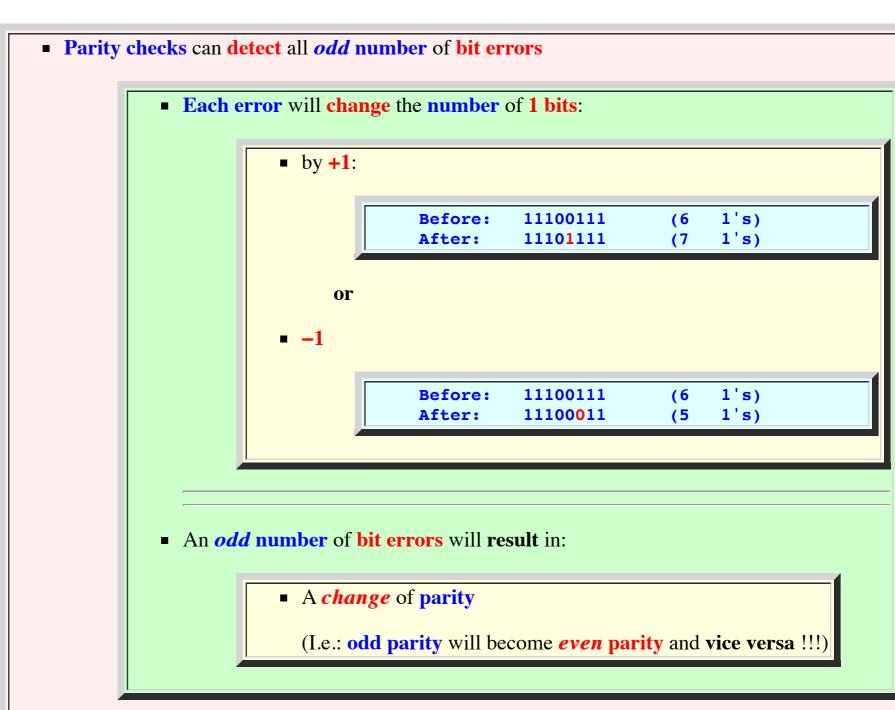
Even parity:	11110000	10101010	11111111	
Odd parity:	11110001	10101011	11111110	

### • Error Checking:

• When the **received message** does not have an **even** (or **odd number of 1's** in the **even** (**or odd**) **parity method**, the **receiver** will **assume** that the **message** is **corrupted** 

### • Property of the *one-dimensional* parity scheme

### • Properties:



### **Example:**

Parity checks can not detect an even number of bit errors

# **Example:**

```
Original data (unprotected): 1111000

Even parity: 11110000

Data with 2 bit errors: 11010100

===> 2 bit errors results in even parity !!!

Error cannot be detected !!!
```

# Multi-dimensional parity-based error detection schemes

- Two-dimensional parity
  - Two-dimensional Parity scheme
    - 2-dimensional parity scheme:
      - Form a *MxN* matrix of bits
      - Add a (even or odd) parity bit to each row and to each column

• Example:

```
Original data (unprotected): 1111000 1010101 1111111

1. For a matrix of bits:

1111000
1010101
1111111

2. Add parity bits:

11110000
10101010
11111111
10100101
```

• **Properties** of the **2**-dimensional parity scheme:

Transmitted data:

11110000
10101010
11111111
10100101

Received with one bit in error:

The receiver can tell which bit was in error from the parity check !!!

#### **Therefore:**

- The receiver can take the initiative to correct the received message!
- The 2-dimensional parity scheme can detect all 2 bit errors...

but it *cannot* correct the error.

### **Example:**

```
Transmitted data:

11110000
10101010
11111111
10100101

Received with 2 bits in error:

11110000
10111010 <---- odd parity
11011111 <---- odd parity
10100101
^^
||
odd parity
```

The **errors** can be **detected**.

**However**, the receiver cannot correct the error:

■ The **cause** of error is *ambiguous*:

Original data:

```
11110000
           10011010
           1111111
           10100101
Error pattern #1:
           11110000
           10001010
                     <--- odd parity
           11011111
                     <---- odd parity
           10100101
       odd parity
Error pattern #2:
           11110000
           10111010 <---- odd parity
           11101111
                     <---- odd parity
           10100101
       odd parity
```

**Both** cases will **result** in the **same** parity pattern !!!

The receiver cannot tell which of these 2 error cases has occured....

So the receiver can not correct the error

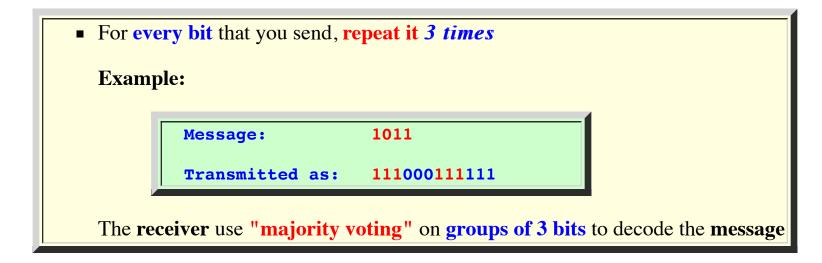
# **Intro to Error** *Correcting* **Codes**

- Prelude: Coding Theory
  - Coding Theory:
    - Coding Theory = a discipline of Mathematics that study the properties of codes and their fitness for specific applications.
  - Use of codes:
    - Data compression
    - Data *encryption*
    - Error correction

Wikipedia: click here

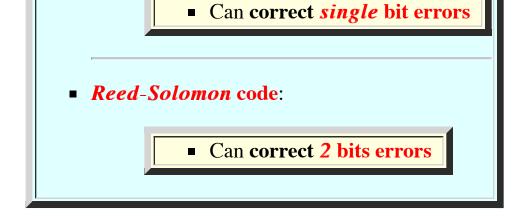
- Error correction codes
  - **Key** to **error correction**:
    - Redundancy....

Example: a naive error correction code



• *More* sophisticated error correction codes:

■ Hamming code:



I will *only* show you *how to* use *Hamming* code

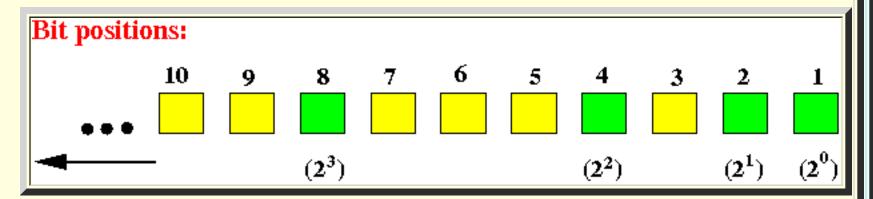
No proofs.... (beyond the scope of this course!)

# Hamming code: encoding procedure

- The Hamming Code
  - Hamming Code:
    - Hamming code is a linear error-correcting code named after its inventor, Richard Hamming.
    - The **Hamming code** can:
      - Detect and correct all 1 bit errors
      - **Detect** *most* 2 bit errors --- but does **not** detect **all** 2 bit errors

(It uses a *minimum* number of extra bits to achieve these properties)

- The *encoding* procedure of the Hamming code
  - **How to encode** data using **Hamming code**:
    - 1. Input = a sequence of bits (0/1 values)
    - 2. **Determine** the **number of** *extra* **bits** to insert (see **example**) **Insert** the **extra bit** *positions* into the **input bits**
    - 3. Compute the values of the extra bit positions
    - 4. Distribute the computed values into the extra bit positions
  - The **Hamming encoding procedure** illustrated by an **example**:
    - 1. **Input** = 101010
    - 2. **Determine** the **number of** *extra* **bits** to insert:
      - The **positions** that corresponds to a **power** of 2 are **Hamming** check bit positions:



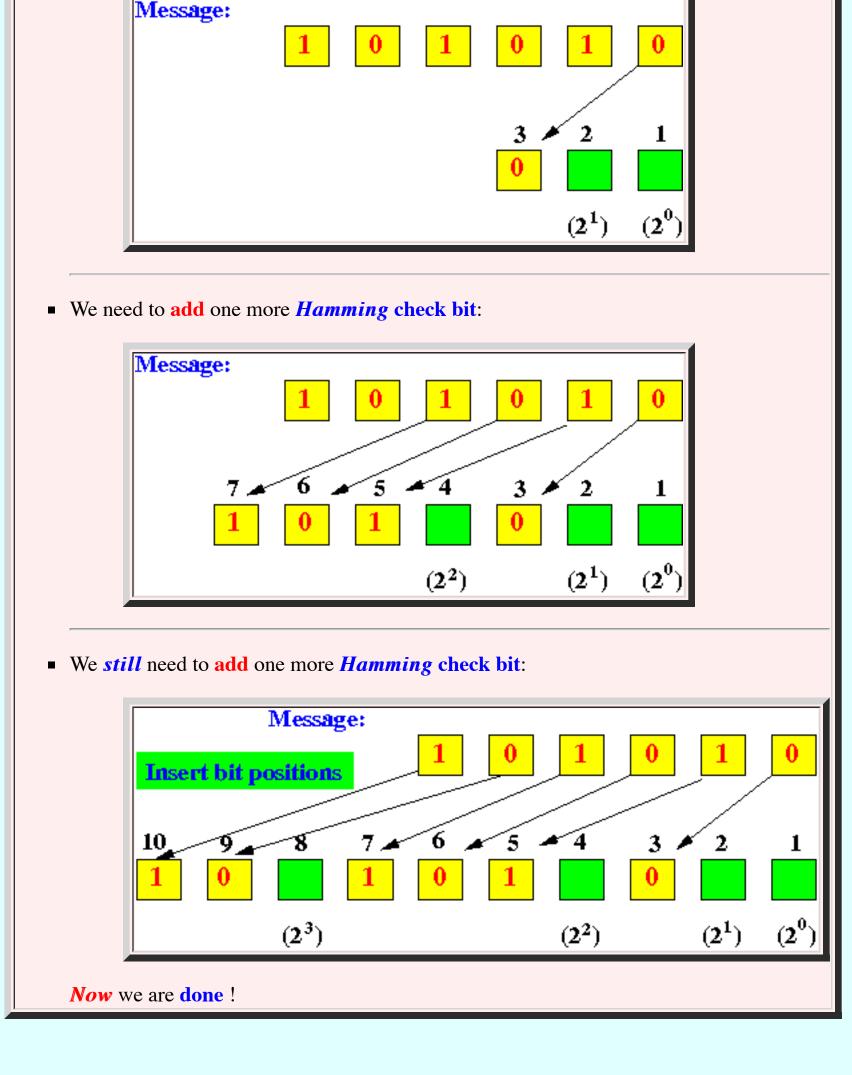
■ *Add* extra bit positions at locations:

**2**<sup>0</sup>, **2**<sup>1</sup>, **2**<sup>2</sup>, ... (and so on)

until all data bits can be accommodated

**Example:** input = **101010** 

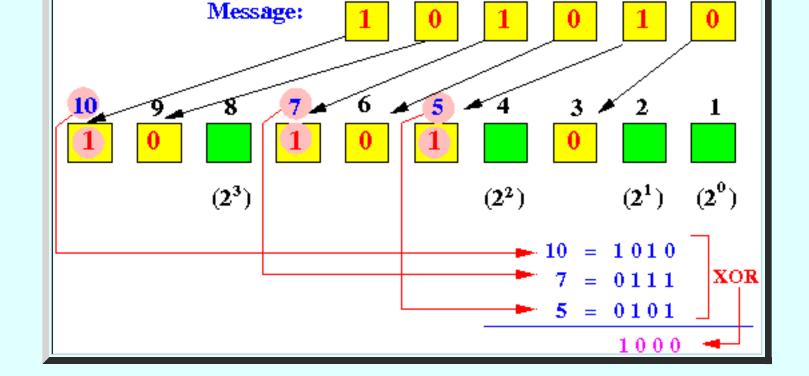
• We start with:



Note:

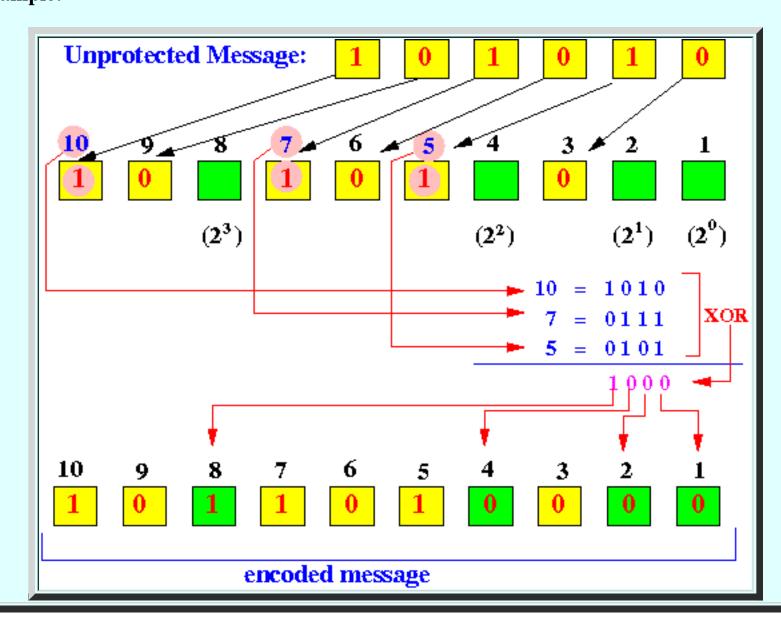
- The *empty* positions are **Hamming check bits** that will be computed next !!!
- 3. *Compute* the values of the bits at the inserted Hamming code positions by:
  - Converting the index of the "1" bits of the input data into a binary number
  - Perform the **XOR operations** column-wise on all the (binary) numbers

**Example:** 



4. Distribute the computed (check) values into the inserted (Hamming) bit positions

# **Example:**



 $\circ$  Summary (of the above):

■ If the **sender** wants to **transmit 101010** using **Hamming Code**, it will **transmit**:

**1011010000** 

# Hamming code: decoding procedure

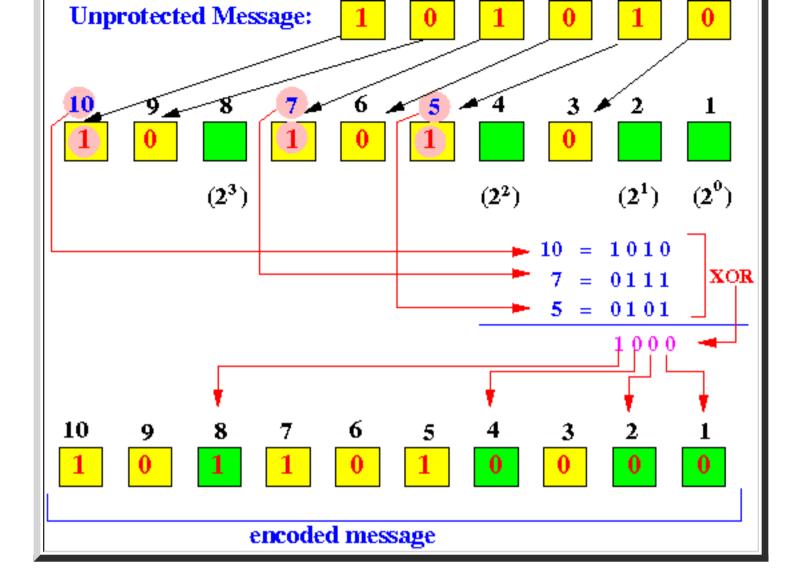
	T 11	TT ·		
	<b>Decoding</b> a	Hamming	COMEN	message
_	Decouning a	TIGHTIIIII	Coucu	HILODOUZE

0	Checking	the	correctness	of a	received	message:
---	----------	-----	-------------	------	----------	----------

- Converting the index of the 1 bit locations into a binary number
- Perform the **XOR operations** on all resulting binary numbers
- If the result = 0 then the messages (most likely) contains no error
  - The decoded message consists of the received bits minus the bits at at positions: 2<sup>0</sup>, 2<sup>1</sup>, 2<sup>2</sup>, ..., 2<sup>k</sup>
- If the result <= length of the message then the messages (most likely) contains 1 error</p>
  - Change the value of the bit at the position given by "result" (computed above)
  - The decoded message consists of the received bits minus the bits at at positions: 2<sup>0</sup>, 2<sup>1</sup>, 2<sup>2</sup>, ..., 2<sup>k</sup>
- If the result > length of the message then the messages (most likely) contains multiple errors
  - The receiver cannot perform correction
  - It will reject the message as corrupted

• Example: **101010** 

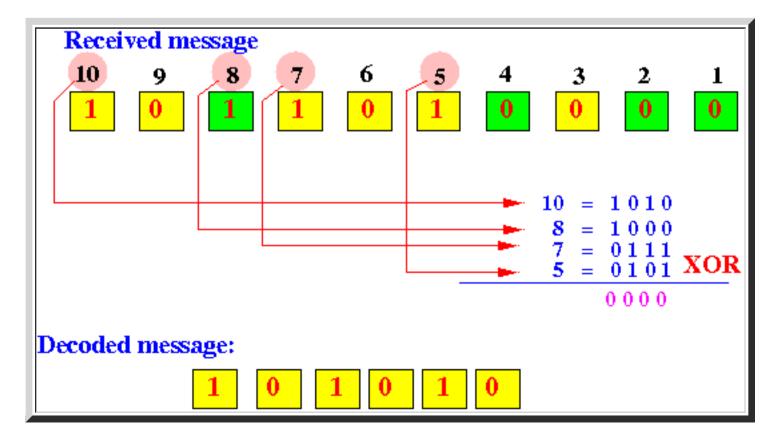
From the discussion above:



we know that the sender will transmit: 1011010000

- Case 1: transmission was received with no error
  - Received message = 1011010000

Outcome of the decode procedure:



**Result:** 

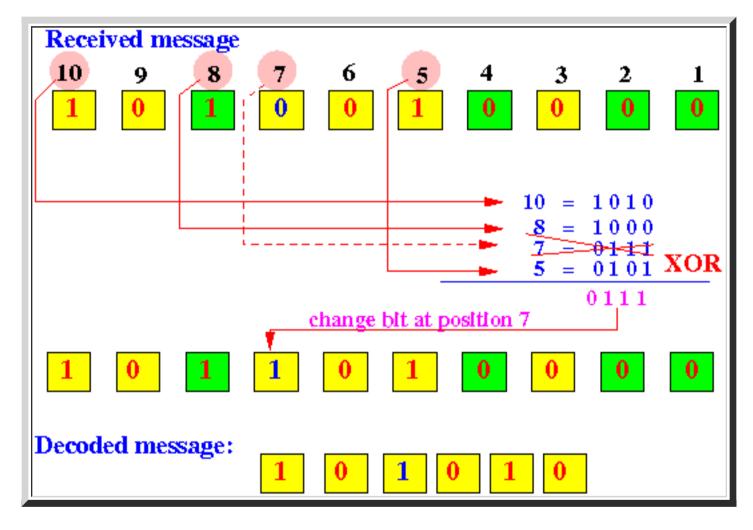


• Case 2: transmission was received with one bit error

■ Received message = 1010010000

(bit position 7 was received in error)

Outcome of the decode procedure:



#### **Result:**

XOR result = 0111 = 7 ( ≤ 10 (message length))
 Conclusion:
 Message has 1 bit error at bit position 7 (which was what happened!)

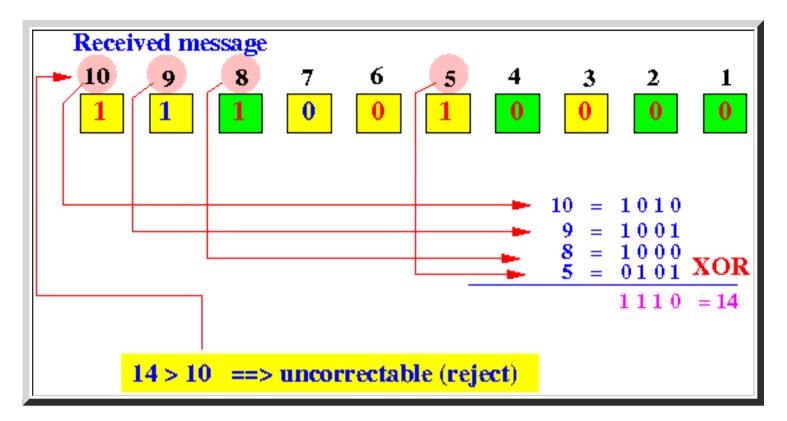
(Try with a bit error in some *other* position!)

Case 3: transmission was received with (detectable) two bits error

■ Received message = 1110010000

(Bit positions 7 and 9 were received in error)

### Outcome of the decode procedure:



### **Result:**

• XOR result = 1110 = 14 ( > 10 (message length))

#### **Conclusion:**

Message has multiple bit errors -- correcting is impossible (which was what happened!)

- Warning: multiple error can cause errorneous correction in Hamming procedure
  - Fact:
- **Some** multiple bit errors will cause the **Hamming decode procedure** to believe that there was a **single bit error**
- Example:
- We use the *same* example:

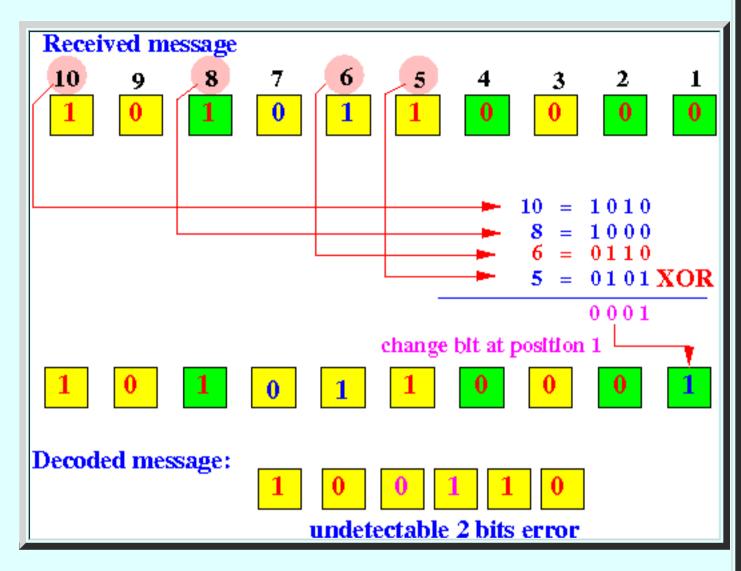
Hamming encode message: 1011010000

(original message was: 101010

■ Suppose bit positions 6 and 7 were received in **error**:

Received message = 1010110000

■ Outcome of the decode procedure:



#### **Result:**

■ XOR result = 0001 = 1 (  $\leq 10$  (message length))

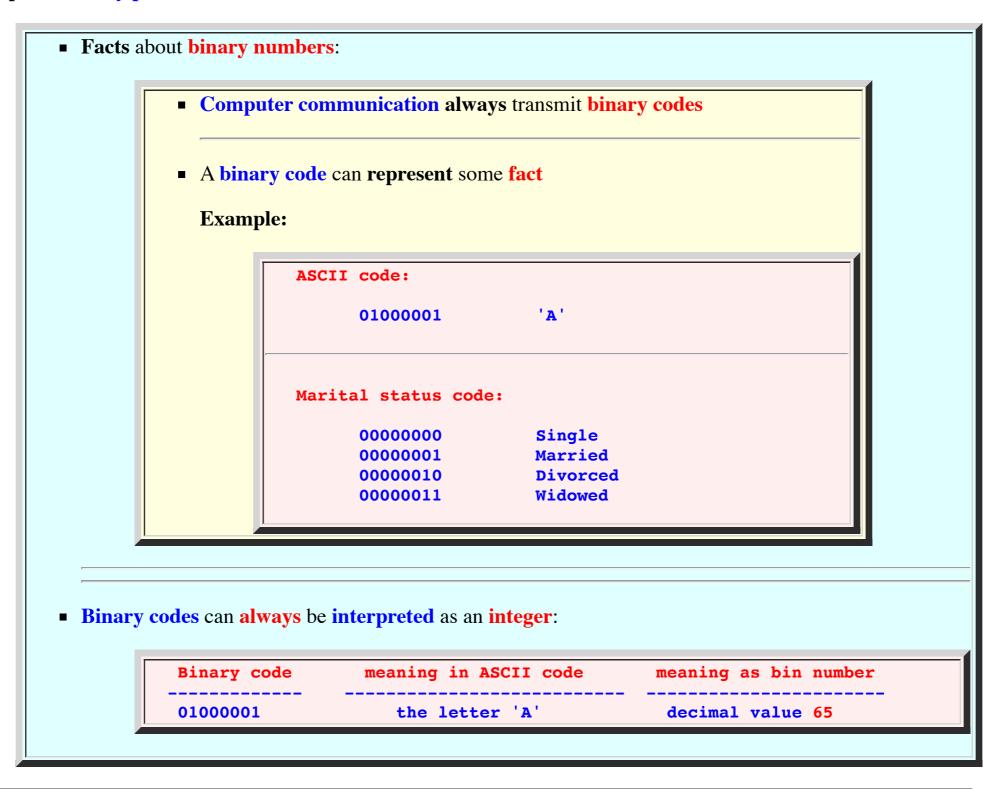
**Conclusion:** 

Messages has 1 bit error at bit position 1...

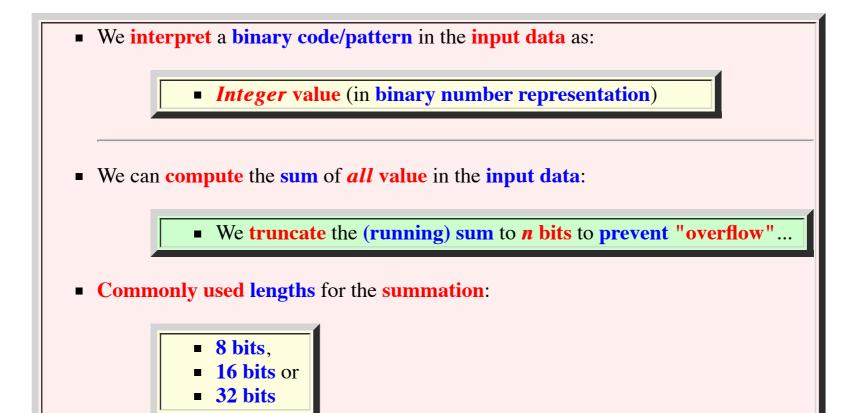
(This is **wrong**!!!)

# Intro to check summing

- Check summing
  - Interpret a binary pattern as numbers:



• Check summing a (input) message:



■ We *append* the (check) sum at the end of the input data

(So the receiver can verify the (check) sum of the *original* input)

### • Example:

- Detecting bit errors with check sums
  - **Checking** for **errors** with **check sum**:
    - The receiver *knows* that the message contains a check sum (byte) at the *end* !!!
    - Check sum verification:
      - 1. The receiver compute the check sum of the received message (excluding the *last* byte)
      - 2. Thereceiver compares the computed check sum against the received check sum
  - Example 1: no bit errors

```
Original message: 01000001 01000010 01000011

Transmitted message: 01000001 01000010 01000011 10000110

Received message: 01000001 01000010 01000011 10000110

Check sum: 01000001 01000010 01000011 10000110
```

• Example 2: *some* bit errors

• Example 3: the check sum method can *miss* detection of some 2 bit errors

- An alternate sum operation: XOR (instead of ADD)
  - Goal of the check sum:
    - The check sum is used to protect the integrity of every (data) value in the entrie message

- The **sum** computed is **meaningless** (i.e., the **sum** is **not** a **total** of some **useful value**)
- A more *efficient* operation used to protect the integrity of serie of values is:



### **Example:**

### Note:

- **XOR** applied to *multiple* input value works like this:
  - If there are an even number of 1's, then XOR output is 0
  - If there are an odd number of 1's, then XOR output is 1
- Property of a message that uses an XOR sum
  - Fact:
- Due to the **property** of the **XOR operation**:
  - The number of 1 bits in *every* column of the XOR check sum computation is *even*

## **Example:**

#### Reason:

- If a column has an odd number of 1 bits:

   XOR will result in 1; making the total # 1 bits in the column even

   If a column has an even number of 1 bits:

   XOR will result in 0; and the total # 1 bits in the column will remain even
- Detecting bit errors with "XOR check sums"
  - **Checking** for **errors** with **check sum**:
    - How to verify a received message:

       The receiver must compute the XOR check sum of the message and the check sum

       If the result = 000...00 (all bits are 0), then:

       The receiver will assume that the message contains no bit errors

       Otherwise:

       The receiver will assume that the message contains some bit errors

### • Example 1: no bit errors case

```
Original message:
                      01000001 01000010 01000011
Transmitted message:
                     01000001
                                         01000011
                                                   01000000
                               01000010
Received message:
                      01000001 01000010 01000011 01000000
Check:
               01000001
               01000010
               01000011
               01000000 XOR
               00000000
"Xor sum" is equal to 00000000 ==> assume no bit errors
```

• Example 3: the XOR sum method can also miss detection of some 2 bit errors

# Pylonimial arithmetic in modulo 2

- CRC: a more *robust* (error detection) check summing technique
  - Cyclic Redundancy Check (CRC):
    - Cyclic Redundancy Check = a check "sum" computation technique based on division of:
      - The message (bits) and
      - A "CRC polynomial"
    - The Cyclic Redundance Check (CRC) method is also known as
      - polynomial code checksum
    - Wikipedia page: click here
  - Facts on CRC:
    - The Cyclic Redundancy Check (CRC) method is the most commonly used "check summing" method in use today
    - When a communication professional use the term "check sum" he/she is referring to a CRC check sum

### Why is **CRC** is **so** *popular*:

- CRC codes can detect common occurring errors that are caused by noise (lightning) in transmission channels.
  - **CRC** can **detect** *n* **consecutive bit errors**
- CRC encoders/decoders are easy to construct:
  - Electronic circuits used to compute the CRC code consists of handful of elementary *gates* and 1-bit memery circuits (D-flipflops)!!!

RC computa	tion
•	
• Fact:	
	■ The CRC check sum computation uses:
	■ Polynomial arithmetic in modulo 2 arithmetic
<ul><li>Furtherr</li></ul>	nora.
o Furtheri	nore.
	■ The polynomials will be represented using bit strings.
	<ul> <li>Polynomial arithmetic operation becomes simple binary number operations us</li> </ul>
	XOR !!!
• We will d	1. Polynomials representation in modulo 2 arithmetic 2. Performing modulo 2 arithmetic on polynomials
	3. Representing polymonials using bit strings
	4. <b>Performing modulo 2 arithmetic</b> on <b>polynomials</b> represented in <b>bit strings</b>
olynomial rep	presentation using modulo 2 arithmetic
_	
• Modulo 2	2 representation:
	■ An even coefficient is mapped to 0
	■ An <i>odd</i> coefficient is mapped to 1
<ul> <li>Example</li> </ul>	representing polynomials in modulo 2:

```
Polynomial:

The polynomial in module 2 representation:

(Coefficient -1 is odd and is mapped to 1)

(Coefficient 2 is even and is mapped to 0)

(Coefficient 1 is odd and is mapped to 1)
```

- Polynomial arithmetic using modulo 2 arithmetic
  - Example: addition

$$(x^3 + x) + (x + 1) = x^3 + 2x + 1$$
  
=  $x^3 + 1$  (mod 2)

• Example: subtraction

$$(x^3 + x) - (x + 1) = x^3 - 1$$
  
=  $x^3 + 1$  (mod 2)

• Example: multiplication

$$(x^{2} + x) \times (x + 1) = (x^{3} + x^{2}) + (x^{2} + x)$$

$$= x^{3} + 2x^{2} + x$$

$$= x^{3} + x \qquad (mod 2)$$

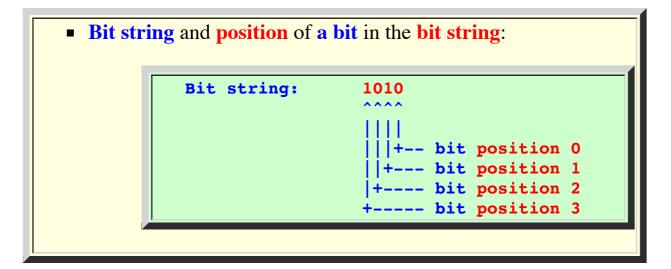
• Example: division

• Modulo 2 arithmetic

• Summary: modulo 2 arithmetic

#### Observation:

- The + (mod 2) operation is the *same* as an *XOR* (exclusive OR) operation used in logic!!!!
- Polynomial representation using a bit string
  - Bit positions



- **Bit string** representation of a **polynomial**:
  - The bit at position k is the *coefficient* of the factor  $x^k$  in the polynomial

### **Example:**

```
Polynomial:

x^5 + x^4 + x^2 + 1

x^5 + x^4 + x^2 + 1

= 1x^5 + 1x^4 + 0x^3 + 1x^2 + 0x^1 + 1x^0

Bit string representation for the polynomial:

110101

0.000

0.000

0.000

0.000

0.000

0.000

0.000

0.000

0.000

0.000

0.000

0.000

0.000

0.000

0.000

0.000

0.000

0.000

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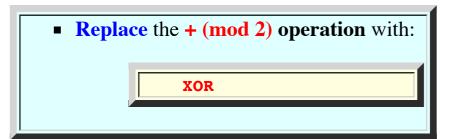
0.000

0.000

0.000
```

```
|||+---- x<sup>2</sup>
||+---- x<sup>3</sup>
|+---- x<sup>4</sup>
+---- x<sup>5</sup>
```

- Polynomial arithmetic using modulo 2 arithmetic with bit strings
  - Polynomial arithmetic in modulo 2 can be perform with bit string representation as follows:



• Example: addition

• Example: subtraction

```
(x^{3} + x) - (x + 1) = x^{3} - 1
= x^{3} + 1  (mod 2)

In bit string representation:
1010 + 0011 ==> 1010
0011
----- XOR
1001 ===> x^{3} + 1 (mod 2)

Same result !!!
```

• Example: multiplication

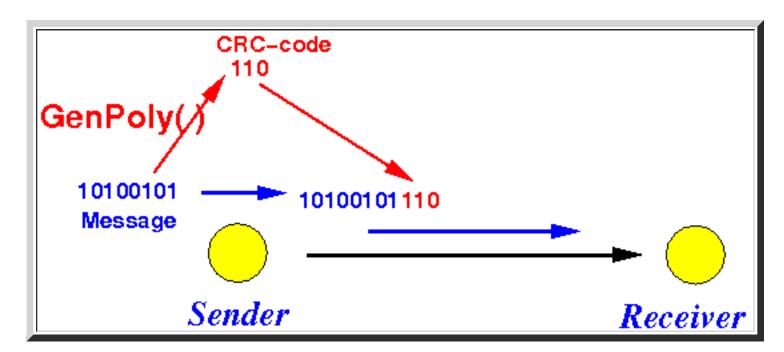
• Example: division

```
x^2 + 1
    x + 1 / x^3 + x^2 + x
               x^3 + x^2
                         x + 1
                             -1
  x^3 + x^2 + x
            -- = x^2 + 1 \qquad remainder \qquad 1 \qquad (mod 2)
   x + 1
In bit string representation:
                           ====> Quotient = x^2 + 1 (mod 2)
                101
         11 / 1110
               11 | |
              ---v
                01
                00
                 10
                 11
                       ===> remainder = 1 \pmod{2}
```

o Conclusion:		
	<ul> <li>We can use the bit string representation and the XOR operation to perform compution for the polynomial arithmetic</li> </ul>	orm the

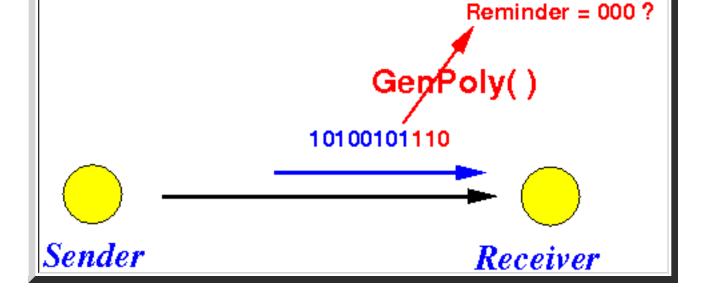
# **CRC** encoding procedure

- Generator polynomial
  - Generator polynomial:
    - Generator polynomial = the divisor polynomial in the polynomial division operation
  - Obvious fact:
    - The sender and the receiver must use the same generator polynomial to encode/decode the messages.
- How to use CRC code
  - **How** the **sender** transmits a **message**:



## **Explanation:**

- The sender *divides* the message by the generator polynomial
- The sender *includes* the *remainder* of the division in the transmitted message
- How the receiver checks a received message for correctness:



## **Explanation:**

- The receiver divides the message + CRC code (remainder) by the same generator polynomial
- The receiver will *assume* there is **no bit error(s)** if the remainder of the division is equal to 0
- Encoding procedure of the Cyclic Redundancy Check (CRC) codes
  - The **CRC** encoding procedure:

```
1. Let:

N = length of the generator polynomial
= number of bits in the generator polynomial

2. Append N-1 ZERO (0) bits to the end of the input message.
```

3. Divide the messages + (N-1) ZEROs by the generator polynomial.

Note:

- The division operation will use the **XOR** operation as the subtract operation
- 4. **Append** the **remainder** of the "division" to the **input message** 
  - The *resulting* message is the CRC *protected* message

• Example: computing the CRC code (= check sum) for a message

```
Generator polynomial: x^3 + x^2 + 1
Genenrator polnomial in bit string: 1101
   (1101 represents the polynomial: 1 \times x^3 + 1 \times x^2 + 0 \times x^1 + 1 \times x^0)
N = 4 (because Generator polynomial has 4 terms, or bits in "1001")
                              x^4 + 1
Input message:
Input message in bit string: 10001
Computing the CRC (check sum) for the message 10001 using Gen Polyn 1101:
   1. Add N-1 (= 3) ZERO's bits to message:
          Message + 3 ZEROs = 10001000
   2. Compute CRC by dividing "message + 3 zeros" by generator polynomial:
                          00011
                  1101 / 10001000
                          1101 <---- **** see Note below !! ****
                          ---- (XOR)
                           1011
                           1101
                            ---- (XOR)
                            1100
                            1101
                             ---- (XOR)
                             000100 < ---- remainder = 100
                    CRC = 100
  3. CRC protected message = 10001100
```

- Note: how to perform the "division"
  - Whenever the *leading* bit of the divident = 1, you get a 1 in the quotient.

```
1
-----
1101 / 10001000 // Even when 1000 < 1101 !!!
1101
---- XOR
0101
```

■ The division uses the **XOR** operation as *subtraction* 

• A longer example....

## • Example:

```
CRC generator polynomial:
                          11011
                                        (N = 5)
Message:
                          11100110
1. Add 4 0-bits to the message:
         11100110 --> 111001100000
2. Divide:
                      10101110
          11011 / 111001100000
                  11011|||
                    1111||
                    0000
                    11111
                    11011
                      1000
                      0000
                      10000
                      11011
                       10110
                       11011
                        11010
                        11011
                         00010
                         00000
```

```
0010

3. CRC protected message = 111001100010
```

- Property of a CRC protected message
  - **Recall**: how is the CRC code computed

```
CRC-polynomial / Original-Message + 000..000
....
CRC-code
```

• A CRC protected message consists of:

```
Original-Message + CRC-code
```

• **Property** of every **CRC** protect message:

- The **remainder** of a **CRC protected message** divided by the **CRC polynomial** (used in the **encoding**) is:
  - Always equal to ZERO !!!

In formula:

```
(Original-Message + CRC-code) % CRC-polynomial == 0 !!!
```

#### **Reason:**

```
CRC-polynomial / Original-Message + 000..000
....
CRC-code
We will have:
```

(The dvision will now bring down the value "CRC-code" which will be subtracted from the remaining value "CRC-code" --- therefore producting the final remainder 0)

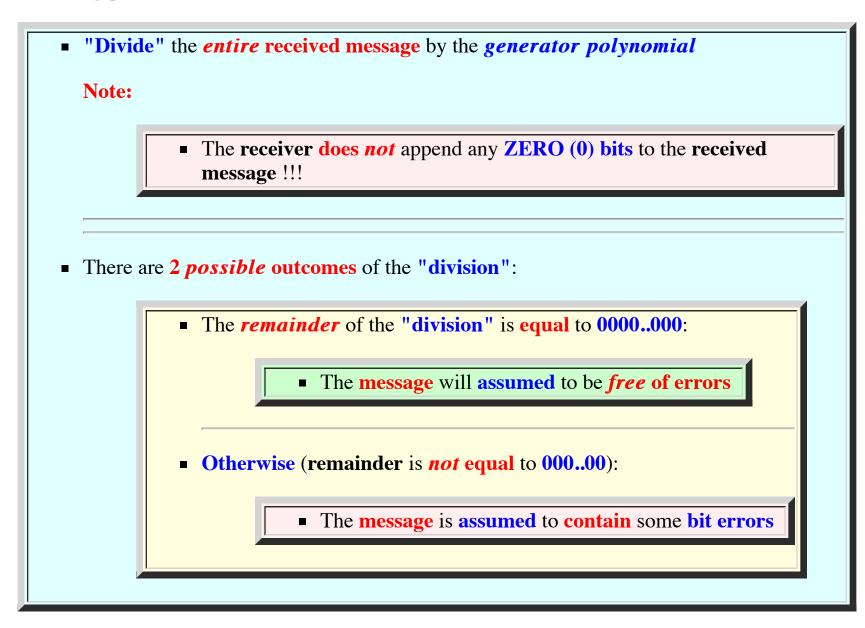
### • Example:

```
Previously, we have computed the CRC code for a message
           CRC generator polynomial:
                                      11011
                                                      (N = 5)
                                      11100110
           Message:
           CRC code =
                                      0010
           CRC protected message = 111001100010
CRC check:
           Divide:
                    Received message 111001100010
                    CRC polynomial
              by
                                       11011
                   (don't care about the quotient)
            11011 / 111001100010
                    11011
                      11111
                      11011
                        10000
                        11011
                         10110
                         11011
                          11011
                          11011
                          000000 <--- Remainder = 0 !!
```



## **CRC** decoding procedure

- Decoding a CRC protected message
  - **CRC** *decoding* procedure:



• Example 1: received message has no errors

```
1101 / 10001100

1101

---- (XOR)

1011

1101

---- (XOR)

1101

1101

---- (XOR)

000 <---- remainder = 000

3. Received message is assumed to be correct

Actual message = 10001 (with the CRC bits removed)
```

• Example 2: received message has two bits in error

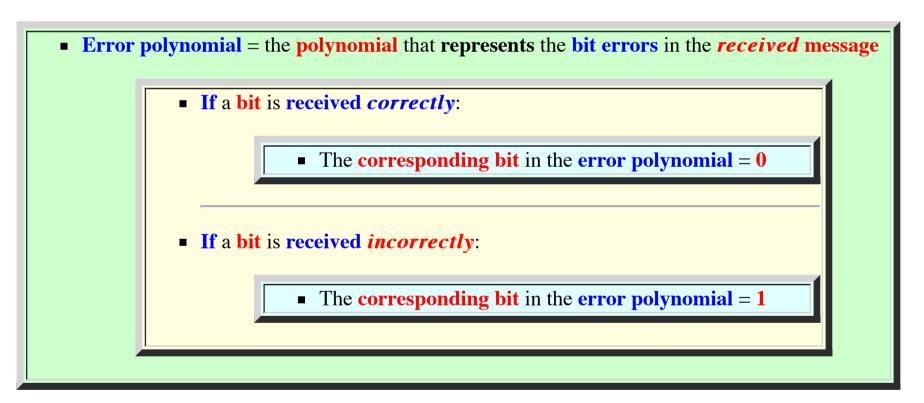
```
Generator polynomial: 1101 (Math. notation: 1 \times x^3 + 1 \times x^2 + 0 \times x^1 + 1 \times x^0)
N = 4 (Generator polynomial has 4 terms, or bits in "1001")
Sender: (from a previous example)
                                       10001
              Message:
              CRC protected message: 10001100
1. Received message = 11101100
2. Check CRC by dividing "received message" by generator polynomial:
                          00011
                  1101 / 11101100
                          1101
                          ---- (XOR)
                           01111
                             1101
                            ---- (XOR)
                              1000
                              1101
                             ---- (XOR)
                              0101 <---- remainder = 101 ≠ 000 !!!
3. Received message is assumed to be corrupted
   Bit errors have been detected !!!
```

# The effect of errors on a CRC message

- Preliminaries
  - **Recall** from the **CRC** decoding procedure:

```
Let M be CRC protected message
Let G be the generator polynomial
If the message M is received with no error, then:
G / M
Remainder = 0000...000
```

- The error polynomial
  - Error polynomial:



• Example:

• **How to determine** the **error polynomial**:

- The CRC computation and the error polynomial
  - Claim:

```
The CRC remainder computed using:

Received Message / generator polynomial

is equal to the CRC remainder computed using:

Error / generator polynomial
```

• Proof:

• Example: (previous example)

■ Message with bit errors:

```
Generator polynomial: 1101 (Math. notation: 1 \times x^3 + 1 \times x^2 + 0 \times x^1 + 1 \times x^0)
N = 4 (Generator polynomial has 4 terms, or bits in "1001")
Sender: (from a previous example)
               Message:
                                         10001
               CRC protected message: 10001100
1. Received message = 11101100
2. Check CRC by dividing "received message" by generator polynomial:
                           00011
                          11101100
                  1101 /
                           1101
                           ---- (XOR)
                            01111
                             1101
                             ---- (XOR)
                                1000
                                1101
```

```
--- (XOR)

0101 <---- remainder = 101 ± 000 !!!

3. Received message is assumed to be corrupted

Bit errors have been detected !!!

We can obtain the same CRC remainder as follows:

1. Received message = 11101100
= 10001100 + 01100000

"error polynomial"

2. Check CRC by dividing the error polynomial by generator polynomial:

1001

1101 / 01100000
1101
----
1000
1101
----
101 <---- same remainder = 101 !!!
```

- When can the CRC method detect error?
  - **Recall** that:

```
    The CRC method has detect (some) bit errors if:
    The remainder of the check procedure ≠ 0000..0000
    In other words:
    if (ReceivedMsg % GenPolynome ≠ 0000...0000) then: bit errors has been detected
```

• From **above**:

```
ReceivedMsg % GenPolynome = Error % GenPolynome !!!
```

### **Therefore:**

```
if ( Error % GenPolynome ≠ 0000...0000 )
then: bit errors has been detected
```

In other words:

```
    If the error polynomial is not divisible by the generator polynomial, then:
    The bit error(s) will be detected by the CRC method
        (Because the remainder of the division is ≠ 000...0000!!!)
    Otherwise:
    The bit error(s) will not be detected by the CRC method
        (Because the remainder of the division is = 000...0000!!!)
```

### • Example:

```
Generator polynomial: 1101 (Math. notation: 1 \times x^3 + 1 \times x^2 + 0 \times x^1 + 1 \times x^0)
N = 4 (Generator polynomial has 4 terms, or bits in "1001")
Sender: (from a previous example)
                                        10001
               Message:
               CRC protected message: 10001100
1. Received message = 11100100
                     = 10001100 XOR 01101000
2. Check CRC by dividing "received message" by generator polynomial:
                          00011
                  1101 / 11100100
                           1101|||
                           ----v||
                            01101
                             1101
                               0000
3. Received message is assumed to be CORRECT
   Bit errors have NOT been detected !!!
```

## Designing good CRC generating polynmials

- Designing CRC *good* polynomials
  - There is a rich Mathematical theory on how to design CRC polynomials with certain desirable properties
    - It's **beyond the scope** of this *Networking* course to go into the **Mathematical details**.
    - I will only show you 2 results
      - 1. How to design a generator polynomial that can detect an even number of bit errors
      - 2. How to design a generator polynomial that can detect *n* consecutive bits in error
- Preliminary: polynomial factoring and primitive polynomial
  - Factors of a polynomial
    - When a polynomial can be obtained by multiplying 2 polynomials:

```
polynomial = polynomial_1 \times polynomial_2
```

we say that:

- The polynomial can be *factored*
- Polynomials polynomial<sub>1</sub> and polynomial<sub>2</sub> are the factors of the polynomial
   polynomial

### **Example:**

```
(x^{2} + x + 1) \times (x + 1) = (x^{3} + x^{2} + x) + (x^{2} + x + 1)
= x^{3} + 2x^{2} + 2x + 1
= x^{3} + 1 \qquad (\text{mod } 2)
Therefore:
x^{2} + x + 1 \quad \text{and} \quad x + 1 \quad \text{are factors of} \quad x^{3} + 1
```

```
because:

x^3 + 1 = (x^2 + x + 1) \times (x + 1)
```

• Primitive polynomial:

■ Primitive polynomial = a polynomial that cannot be factored

I.e., A primitive polynomial cannot be written as a product of 2 different polynomials:

polynomial<sub>1</sub> × polynomial<sub>2</sub>

- Generator polynomial that can detect all *odd* number of bit errors
  - Parity Property:
    - A generator polynomial that contains the factor x+1 (= 11) can detect all errors that affect an odd number of bits
      - This property is called the Parity property

### **Example:**

- Property of polynomials that contains "x+1" (= 11) as a factor
  - **Property** of **polynomial** with (x+1) as a **factor**:
    - Every message that is CRC protected by a polynomial with (x+1) as a factor has this property:
      - The number of 1 bits in the CRC protected message has an even number of 1 bits

### • Example:

```
Message: 10011 (has an odd # 1 bits)

CRC polynomial: 1001 (1001 has 11 as factor)

Compute CRC code:

1001 / 10011000
1001
----
01000
1001
-----
001 <----- Remainder

CRC protected message:

10011001 (has an even number 1 bits !!!)
```

- Why a polynomial containing "x+1" (= 11) can detect an even number of bit errors
  - Why a polynomial containing "x+1" (= 11) can detect an *odd* number of bit errors:
    - An *odd* number of bit errors will result in a *received* message containing an *odd* number of 1 bits

#### **Reason:**

- The CRC protected message has an *even* number of 1 bits (see previous claim)
- There are **2 possible bit errors**

```
Bit error type | Effect

O bit sent --> 1 bit received | # 1 bits increased by 1

1 bit sent --> 0 bit received | # 1 bits decreased by 1
```

- If there are 0 bit errors: the received message has an even number of 1 bits
   If there is 1 bit error: the received message has an odd number of 1 bits
   If there are 2 bit errors: the received message has an even number of 1 bits
   If there are 3 bit errors: the received message has an odd number of 1 bits
   And so on ....
- Fact:
- When a received message containing an odd number of 1 bits is divided by the genertor polynomial:
  - The remainder of the division will *not* be equal to 000...000!!!

#### This makes sense, because:

A correctly received CRC protected message will always contain an even number of 1 bits!!

Therefore:

- A received message with an *odd* number of 1 bits can *never* be *assumed* correct !!!
- Generator polynomial that can detect *k* consecutive bit errors
  - **Property** of a **primitive polynomial** of **degree** *n*:
    - If  $x^n + \dots + 1$  is a primitive polynomial of degree n, then:
      - The CRC code can detect any *n* or less *consecutive* bit errors

• Reason:

### • Therefore:

■ A generator polynomial containing an primitive polynomial of degree n can detect k consecutive bit errors ---- for any  $k \le n$ ).

- Commonly used (international standard) generator polynomials
  - Some commonly used CRC generator polynomials:

Name	Generator polynomial
CRC-CCITT	10001000000100001
CRC-16	1100000000000101
CRC-32	100000100110000010001110110110111

• More CRC polynomials at Wikipedia page: click here

# Implementing the CRC scheme in hardware

- Implementing the CRC division algorithm with hardware
  - An important advantage of CRC codes is:
    - The "divide" algorithm used in the CRC can be easily implemented with cheap hardware:
       Shift-register (actually: D-flipflops) and
       XOR circuits
- Shift registers
  - Shift-register circuit:
    - 1 bit shift register = a circuit with that copies the input value to its output during the time that the clock signal changes from  $0 \Rightarrow 1$  (or from  $1 \Rightarrow 0$ )
  - You can see the **behavior** of a **serie** of **shift-register** in this demo:

/home/cs455001/demo/Logic-Sim/shift-register

- XOR circuit
  - XOR circuit:
    - XOR circuit = a circuit with 2 inputs and one output

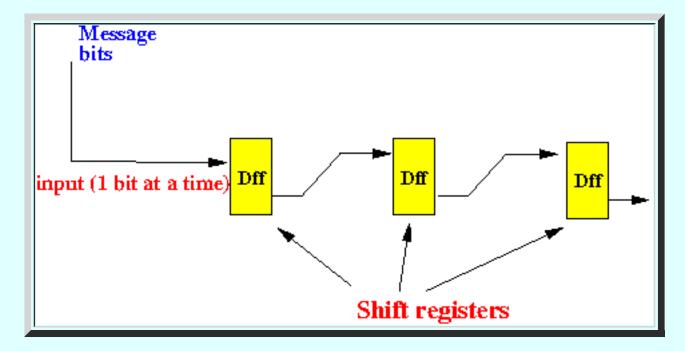
      The XOR function is as follows:

      O XOR O = O
      O XOR 1 = 1
      1 XOR 0 = 1
      1 XOR 1 = 0
  - You can see the **behavior** of the **XOR** circuit in this demo:

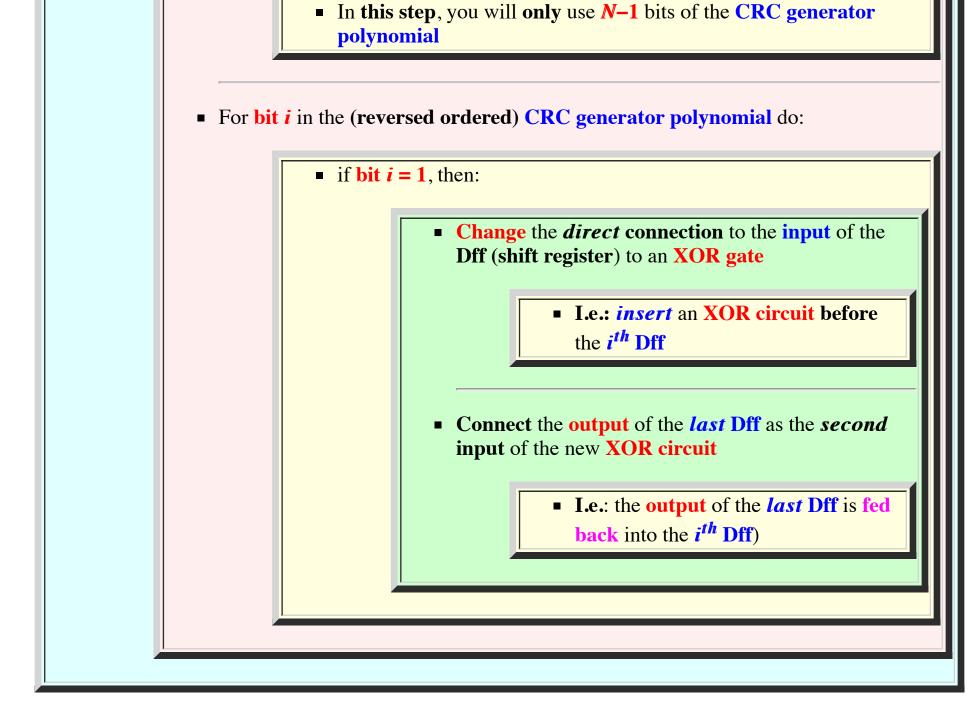
/home/cs455001/demo/Logic-Sim/xor

- Hardware CRC implementation
  - How to create a circuit that compute the CRC code for a given generator polynomial:
    - Let
- N =length of the generator polynomial
- Use (*N*-1) **D-flipflops** (= shift registers) to store the remainder of the division.
- Start by
  - Putting (N-1) Dffs in a row.
     Connect the output of the i<sup>th</sup> Dff to the input of the i+1<sup>th</sup> Dff
     Example: (N = 4 or N-1 = 3)

    Message bits
    Dff
    Dff
    Dff
    Dff
    Dff
    Shift registers
- The input of the first Dff will receive the message *one* bit at a time:

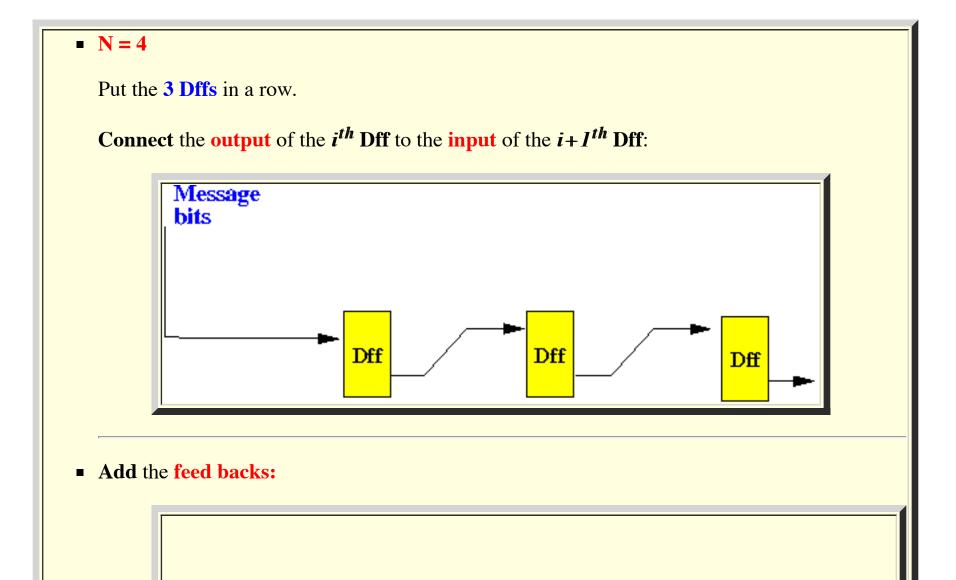


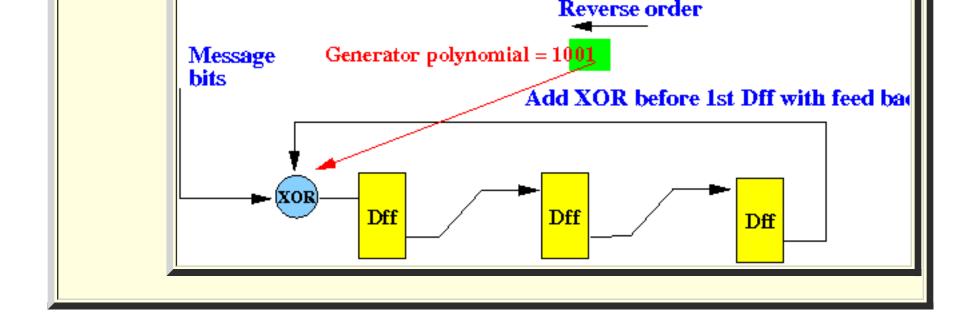
- We must **modify** the **circuit** as follows:
  - Read the **CRC generator polynomial** in the *reverse* order.



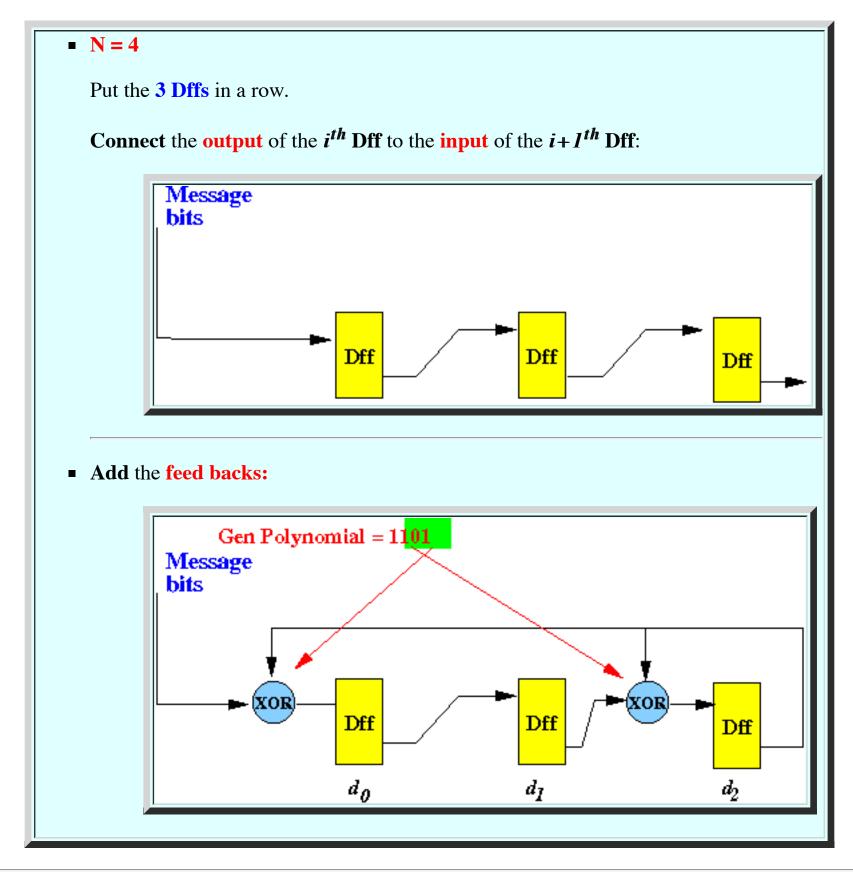
It's much easier to see and example (I'll do it in class)

• Example: compute circuit for generator polynomial 1001





• Example: compute circuit for generator polynomial 1101



- Computing the CRC with the hardware implementation
  - **How to** compute the **CRC** for an **input message**:

Reset all the Dffs (memory elements) to ZERO ■ Toggle key 'a' to ONE to clear the shift registers ■ Toggle key 'a' back to **ZERO** (blank cell) so you can run the circuit Present the first bit of the message ■ Use **key 0** to set it to the **value** of the **first bit** (**0** or **1**) • The **input bits** are presented from **left to right**: Message: 101010 Present the bits as follows: • Give the compute circuit a clock signal • You must **toggle** the key **1** *twice* Present the second bit of the message (see above) • Give the **compute circuit** another **clock signal** (see above) • And so on, until all bits in the message are processed Note: ■ Remember that the sender must *append N*-1 ZERO (0) bits to the input message ■ These **N-1 ZERO** (0) bits must be **processed** *also*!! • After processing all bits: Sender: **Remainder** of the "division" = values of the shift registers read in Receiver: ■ If remainder of the "division" = all 0's, then *accept* the message • Otherwise, *reject* the message

- Example Program: (Circuit for generator polynomial 1101)
  - Prog file: <u>click here</u>

#### How to use it:

- Run: /home/cs455001/demo/Logic-Sim/CRC1
- Make  $\mathbf{a} = \mathbf{0}$  (blank) to run circuit
- Toggle **0:Input** to the next bit in input
- Press 1:Clk to give circuit a clock signal (2 toggles per input bit).

This is a **logic-sim** program from my **CS355** class...

### **Example:**

```
00011
------
1101 / 10001000
1101
---- (XOR)
1011
1101
---- (XOR)
1100
1101
---- (XOR)
000100 <---- remainder = 100
```

#### Do this:

- Toggle 2 to run circuit
- Toggle 0 (input) to 1, then toggle 1 (clock twice)
- Toggle 0 (input) to 0, then toggle 1 (clock **twice**)
- Toggle 0 (input) to 0, then toggle 1 (clock **twice**)
- Toggle 0 (input) to **0**, then toggle 1 (clock **twice**)
- Toggle 0 (input) to 1, then toggle 1 (clock twice)
- Toggle 0 (input) to 0, then toggle 1 (clock **twice**)
- Toggle 0 (input) to 0, then toggle 1 (clock **twice**)
- Toggle 0 (input) to **0**, then toggle 1 (clock **twice**)
- Read the bits backwards: 100