

Chapter 5

Data Link Layer

Reti di Elaboratori

Corso di Laurea in Informatica

Università degli Studi di Roma "La Sapienza"

Canale A-L

Prof.ssa Chiara Petrioli

Parte di queste slide sono state prese dal materiale associato al libro
Computer Networking: A Top Down Approach, 5th edition.

All material copyright 1996-2009

J.F Kurose and K.W. Ross, All Rights Reserved

Thanks also to Antonio Capone, Politecnico di Milano, Giuseppe Bianchi and
Francesco LoPresti, Un. di Roma Tor Vergata

Chapter 5: The Data Link Layer

Our goals:

- understand principles behind data link layer services:
 - error detection, correction
 - sharing a broadcast channel: multiple access
 - link layer addressing
 - reliable data transfer, flow control: *done!*
- instantiation and implementation of various link layer technologies

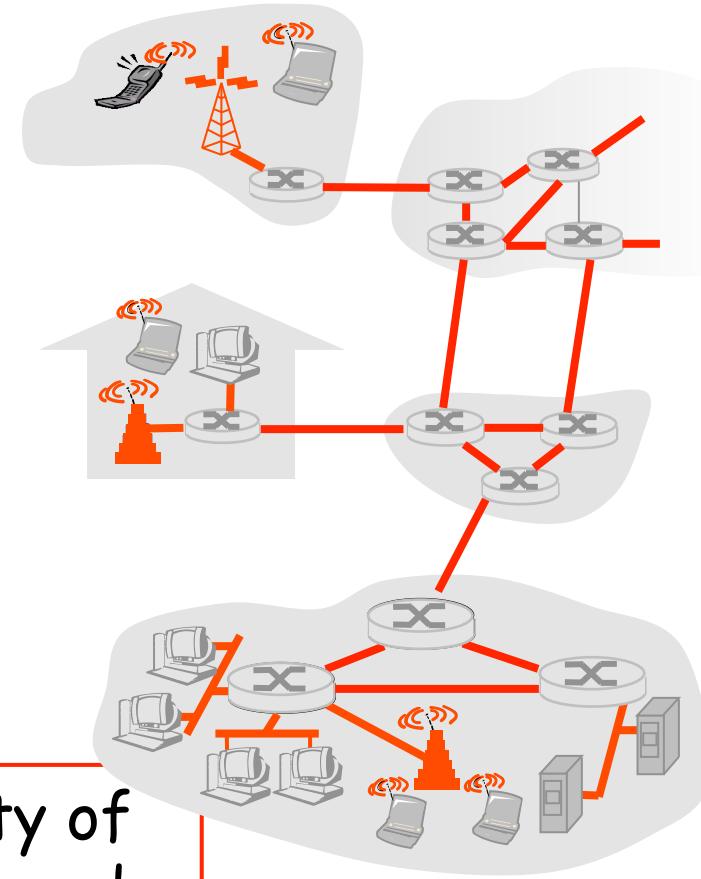
Link Layer

- 5.1 Introduction and services
- 5.2 Error detection and correction
- 5.3 Multiple access protocols
- 5.4 Link-layer Addressing
- 5.5 Ethernet
- 5.6 Link-layer switches
- 5.7 PPP
- 5.8 Link virtualization: MPLS
- 5.9 A day in the life of a web request

Link Layer: Introduction

Some terminology:

- hosts and routers are **nodes**
- communication channels that connect adjacent nodes along communication path are **links**
 - wired links
 - wireless links
 - LANs
- layer-2 packet is a **frame**, encapsulates datagram



data-link layer has responsibility of transferring datagram from one node to adjacent node over a link

Link layer: context

- datagram transferred by different link protocols over different links:
 - e.g., Ethernet on first link, frame relay on intermediate links, 802.11 on last link
 - each link protocol provides different services
 - e.g., may or may not provide rdt over link
- transportation analogy
- trip from Princeton to Lausanne
 - limo: Princeton to JFK
 - plane: JFK to Geneva
 - train: Geneva to Lausanne
 - tourist = **datagram**
 - transport segment = **communication link**
 - transportation mode = **link layer protocol**
 - travel agent = **routing algorithm**

Link Layer Services

- PHY layer accepts only a raw bit stream and attempts to deliver to destination

0110001100001100000001001100000100001

- Communication is not necessarily error free
- Multiplexing of different flows of information
 - Data link layer breaks the bit stream up into discrete frames (FRAMING) and computes the checksum for each frame (ERROR DETECTION)

Link Layer Services

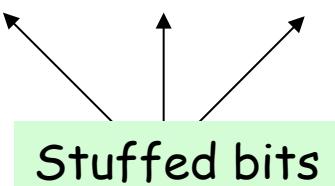
Framing:

- encapsulate datagram into frame, adding header, trailer
- How to delimit frames:
 - We cannot count on some time gap (strong synch requirement and jitter requirement)
 - Character count: A field in the header specifies the number of characters in the frame (OK but loose synch in case of transmission error)
 - Starting and ending characters with character stuffing
 - ES ASCII character sequence DLE STX (Data Link Escape Start of TeXt)...DLE ETX (ETX=End of TeXt)
 - What if binary data are transmitted with sequences corresponding to DLE STX or SLE ETX occurring in the data?
 - Character stuffing: before transmitting add DLE before each of
 - such sequences in the data: DLE STX → DLE DLE STX

Link Layer Services

Framing:

- encapsulate datagram into frame, adding header, trailer
- How to delimit frames:
 - Starting and ending flags with bit stuffing
 - Each frame begins and ends with a special bit pattern, e.g. 01111110 (flag sequence)
 - Techniques to avoid problems in case the flag sequence appears in data: whenever data link layer encounters five consecutive ones in the data add a 0 bit in the outgoing bit stream (removed at the other end of the link) → bit stuffing
 - Es.: (a) 011011111111111111110010
 - (b) 011011111011111011111010010



Link Layer Services

Framing:

- encapsulate datagram into frame, adding header, trailer
- How to delimit frames:
 - Physical layer coding variations
 - For instance if Manchester encoding used a High-High or Low-Low sequence
 - A combination of character count and one of the other typically used

Link Layer Services

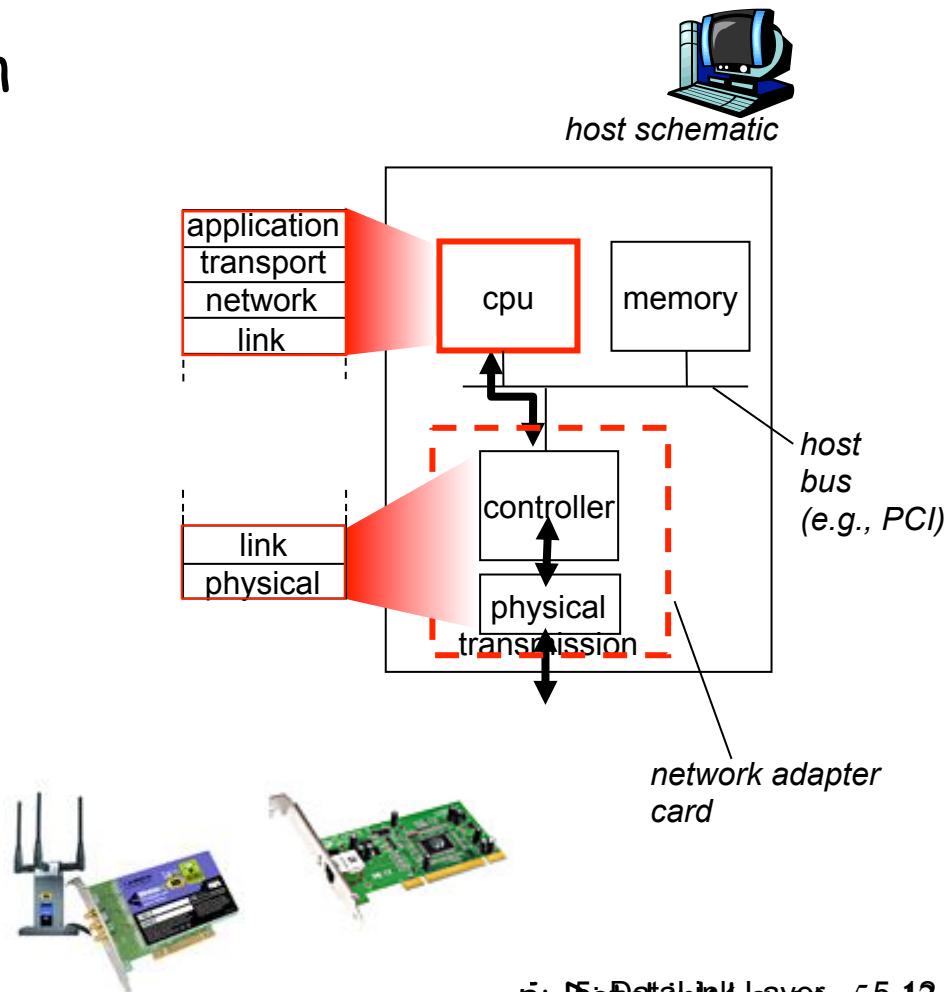
- *link access*
 - channel access if shared medium
 - avoids or limits the effect of collisions over a broadcast channel
- *addressing*
 - “MAC” addresses used in frame headers to identify source, dest
 - different from IP address!
- *error detection:*
 - errors caused by signal attenuation, noise.
 - receiver detects presence of errors:
 - signals sender for retransmission or drops frame
- *error correction:*
 - receiver identifies *and corrects* bit error(s) without resorting to retransmission
- *half-duplex and full-duplex*
 - with half duplex, nodes at both ends of link can transmit, but not at same time

Link Layer Services (more)

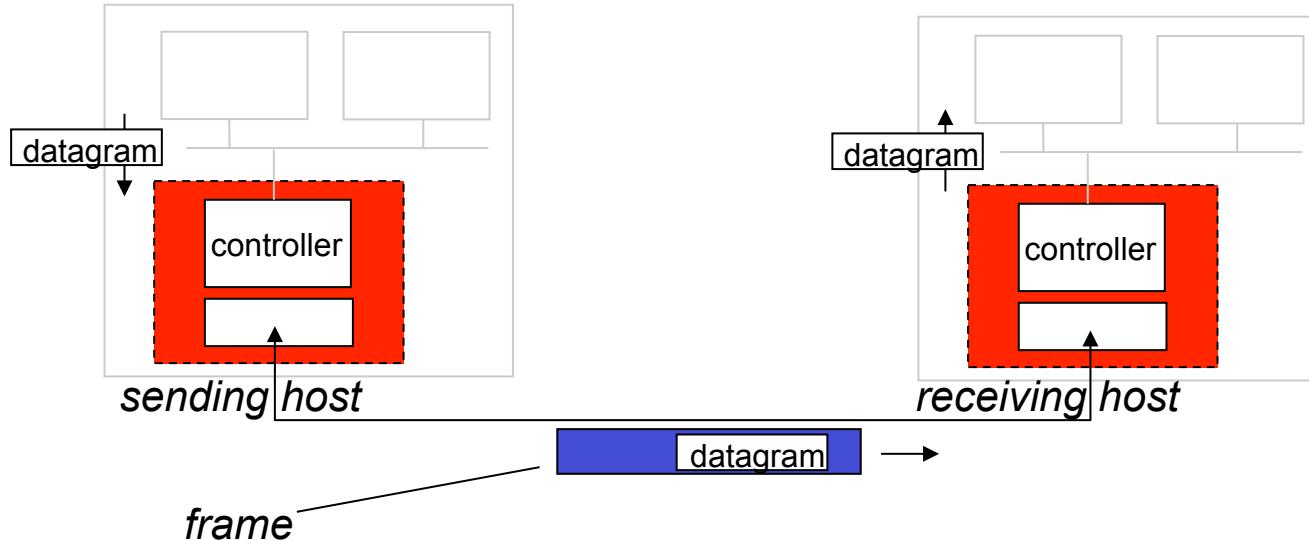
- *reliable delivery between adjacent nodes*
 - we learned how to do this already (chapter 3)!
 - seldom used on low bit-error link (fiber, some twisted pair)
 - wireless links: high error rates
 - Q: why both link-level and end-end reliability?
- *flow control:*
 - pacing between adjacent sending and receiving nodes

Where is the link layer implemented?

- in each and every host
- link layer implemented in “adaptor” (aka *network interface card NIC*)
 - Ethernet card, PCMCIA card, 802.11 card
 - implements link, physical layer
- attaches into host’s system buses
- combination of hardware, software, firmware



Adaptors Communicating



- **sending side:**
 - encapsulates datagram in frame
 - adds error checking bits, rdt, flow control, etc.
- **receiving side**
 - looks for errors, rdt, flow control, etc
 - extracts datagram, passes to upper layer at receiving side

Link Layer

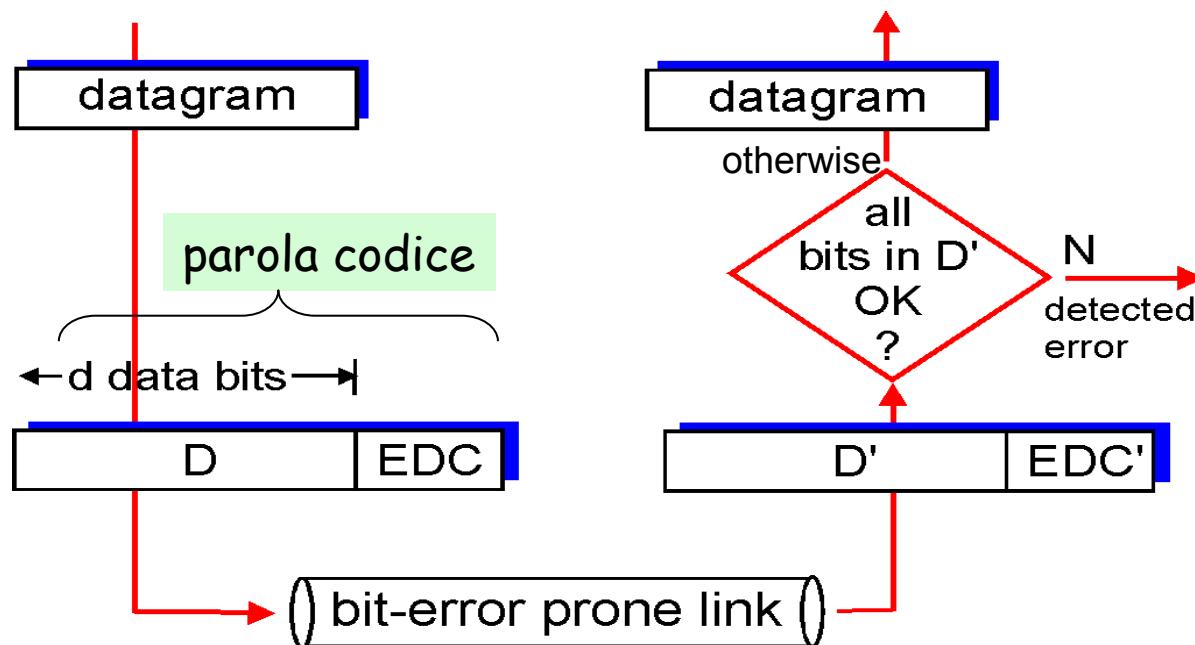
- 5.1 Introduction and services
- 5.2 Error detection and correction
- 5.3 Multiple access protocols
- 5.4 Link-layer Addressing
- 5.5 Ethernet
- 5.6 Link-layer switches
- 5.7 PPP
- 5.8 Link virtualization: MPLS
- 5.9 A day in the life of a web request

Error Detection

EDC= Error Detection and Correction bits (redundancy)

D = Data protected by error checking, may include header fields

- Error detection not 100% reliable!
 - protocol may miss some errors, but rarely
 - larger EDC field yields better detection and correction



Distanza di Hamming

- Date due parole codice e.g., 10001001 e 10110001 è possibile determinare in quanti bit 'differiscono' (XOR delle due parole e contate il numero di 1 del risultato)
 - Il numero di posizioni nelle quali le due parole di codice differiscono determina la loro distanza di Hamming
 - Se due parole codice hanno una distanza di Hamming d ci vorranno d errori sui singoli bit per tramutare una parola di codice nell'altra
 - Per come sono usati i bit di ridondanza se la lunghezza delle parole di codice è $n=m+r$ sono possibili 2^m messaggi dati ma non tutte le 2^n parole codice
 - la distanza di Hamming di un codice è la minima distanza di Hamming tra due parole codice

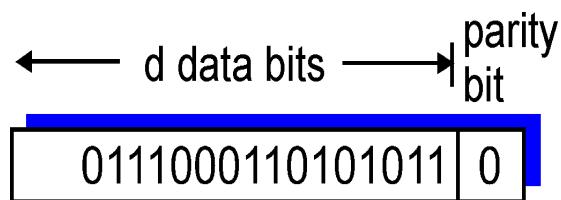
Distanza di Hamming

- Date due parole codice e.g., 10001001 e 10110001 è possibile determinare in quanti bit 'differiscono' (XOR delle due parole e contate il numero di 1 del risultato)
 - Per come sono usati i bit di ridondanza se la lunghezza delle parole di codice è $n=m+r$ sono possibili 2^m messaggi dati ma non tutti 2^n parole codice
 - la distanza di Hamming di un codice è la minima distanza di Hamming tra due parole codice
 - Per fare il detection di d errori serve un codice con distanza di Hamming $d+1$
 - Per correggere d errori serve un codice con distanza di Hamming $2d+1$

Parity Checking

Single Bit Parity:

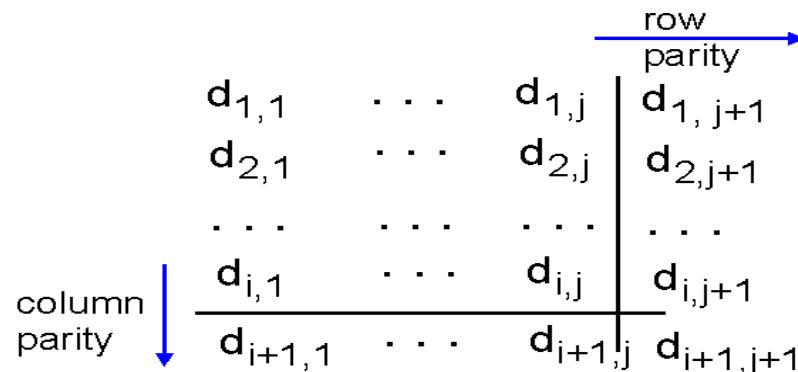
Detect single bit errors



Schema di parità dispari:
Il mittente include un bit addizionale e sceglie il suo valore in modo che il numero di uno nei $d+1$ bit sia dispari

Two Dimensional Bit Parity:

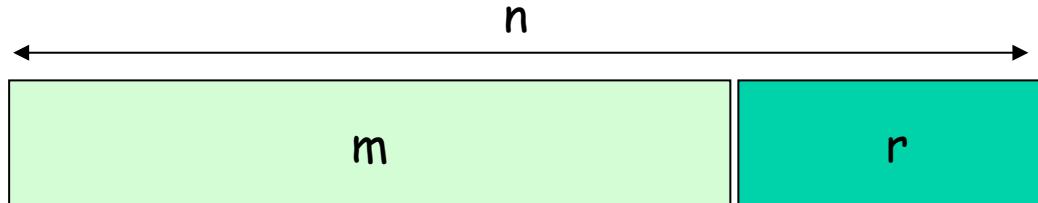
Detect and correct single bit errors



101011
111100
011101
001010
no errors

101011
101100
011101
001010
parity error
correctable single bit error

Quanta ridondanza serve per correggere errori singoli?

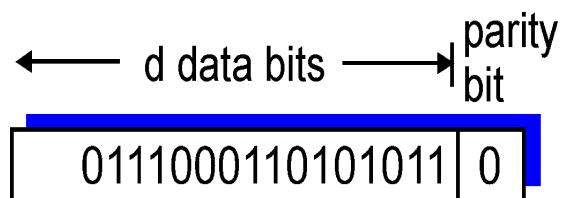


- 2^m messaggi legali
 - Ciascuna parola codice legale ne ha n a distanza 1
 - Ciascuno dei 2^m messaggi legali deve avere $(n+1)$ sequenze di bit a lui associate
 - $(n+1)2^m \leq 2^n$
 - $n=m+r$
 - $(m+r+1) \leq 2^r$
- Lower bound su r

Parity Checking

Single Bit Parity:

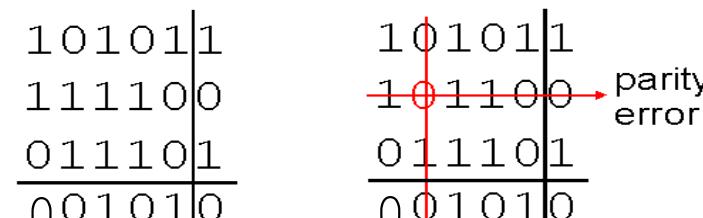
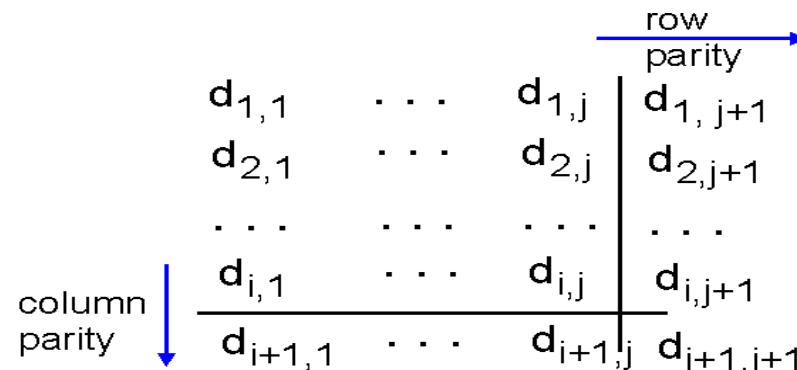
Detect single bit errors



Schema di parità dispari:
Il mittente include un bit addizionale e sceglie il suo valore in modo che il numero di uno nei $d+1$ bit sia dispari

Two Dimensional Bit Parity:

Detect and correct single bit errors



Schemi semplici possono essere sufficienti nel caso di errori casuali
Cosa si può fare nel caso di errori a burst?

- Maggiore ridondanza
- Interleaving

Internet checksum (review)

Goal: detect “errors” (e.g., flipped bits) in transmitted packet (note: used at transport layer only)

Sender:

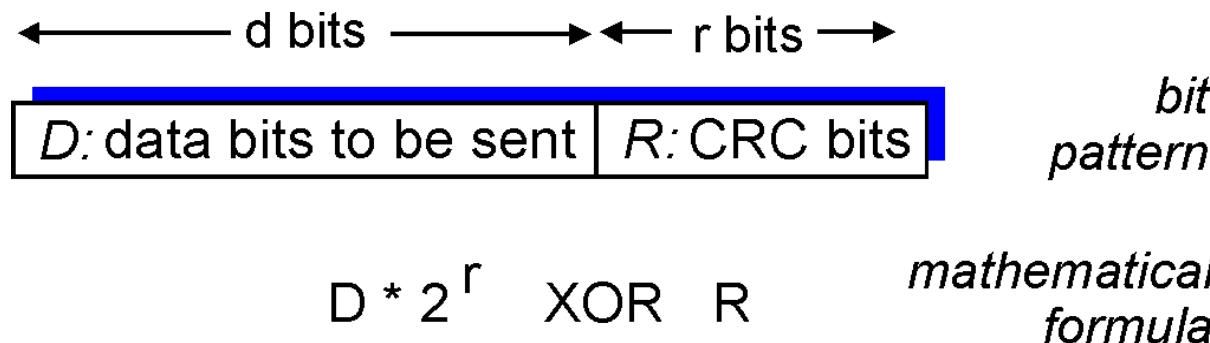
- treat segment contents as sequence of 16-bit integers
- checksum: addition (1's complement sum) of segment contents
- sender puts checksum value into UDP checksum field

Receiver:

- compute checksum of received segment
- check if computed checksum equals checksum field value:
 - NO - error detected
 - YES - no error detected.
But maybe errors nonetheless?

Checksumming: Cyclic Redundancy Check

- view data bits, D , as a binary number
- choose $r+1$ bit pattern (generator), G
- goal: choose r CRC bits, R , such that
 - $\langle D, R \rangle$ exactly divisible by G (modulo 2)
 - receiver knows G , divides $\langle D', R' \rangle$ by G . If non-zero remainder: error detected!
 - can detect all burst errors less than $r+1$ bits
- widely used in practice (Ethernet, 802.11 WiFi, ATM)



CRC

- r è l'ordine del polinomio generatore $G(x)$
- Appendi r bit zero al messaggio $M(x)$ che ora corrisponde a $x^r M(x)$
- dividi $x^r M(x)$ per $G(x)$ modulo 2
- Sottrai (modulo 2) il resto della divisione da $x^r M(x) \rightarrow$ si ottiene $T(x)$, il risultato da trasmettere

- In ricezione controlla che il resto della divisione per $G(x)$ sia 0
- Estrai la parte di messaggio $M(x)$

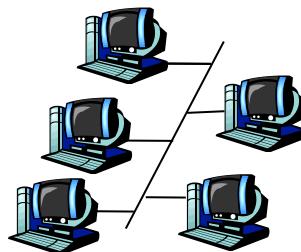
Link Layer

- 5.1 Introduction and services
- 5.2 Error detection and correction
- 5.3 Multiple access protocols
- 5.4 Link-layer Addressing
- 5.5 Ethernet
- 5.6 Link-layer switches
- 5.7 PPP
- 5.8 Link virtualization: MPLS
- 5.9 A day in the life of a web request

Multiple Access Links and Protocols

Two types of “links”:

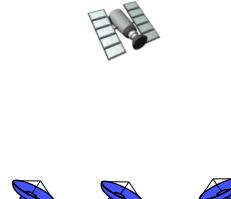
- point-to-point
 - PPP for dial-up access
 - point-to-point link between Ethernet switch and host
- broadcast (shared wire or medium)
 - old-fashioned Ethernet
 - upstream HFC
 - 802.11 wireless LAN



shared wire (e.g.,
cabled Ethernet)



shared RF
(e.g., 802.11 WiFi)



shared RF
(satellite)



humans at a
cocktail party
(shared air, acoustical)

Multiple Access protocols

- single shared broadcast channel
- two or more simultaneous transmissions by nodes:
interference
 - **collision** if node receives two or more signals at the same time

multiple access protocol

- distributed algorithm that determines how nodes share channel, i.e., determine when node can transmit
- communication about channel sharing must use channel itself!
 - no out-of-band channel for coordination

Ideal Multiple Access Protocol

Broadcast channel of rate R bps

1. when one node wants to transmit, it can send at rate R.
2. when M nodes want to transmit, each can send at average rate R/M
3. fully decentralized:
 - no special node to coordinate transmissions
 - no synchronization of clocks, slots
4. simple

MAC Protocols: a taxonomy

Three broad classes:

Channel Partitioning

- divide channel into smaller “pieces” (time slots, frequency, code)
- allocate piece to node for exclusive use

Random Access

- channel not divided, allow collisions
- “recover” from collisions

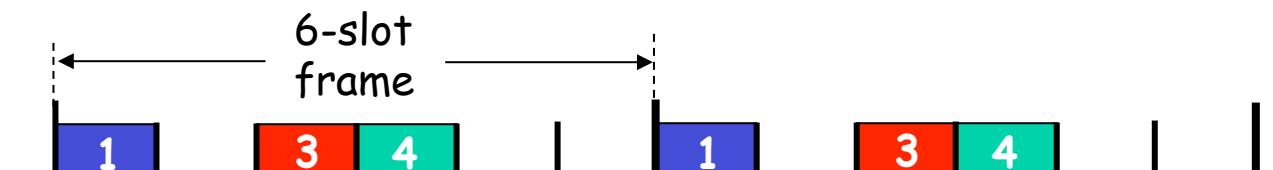
“Taking turns”

- nodes take turns, but nodes with more to send can take longer turns

Channel Partitioning MAC protocols: TDMA

TDMA: time division multiple access

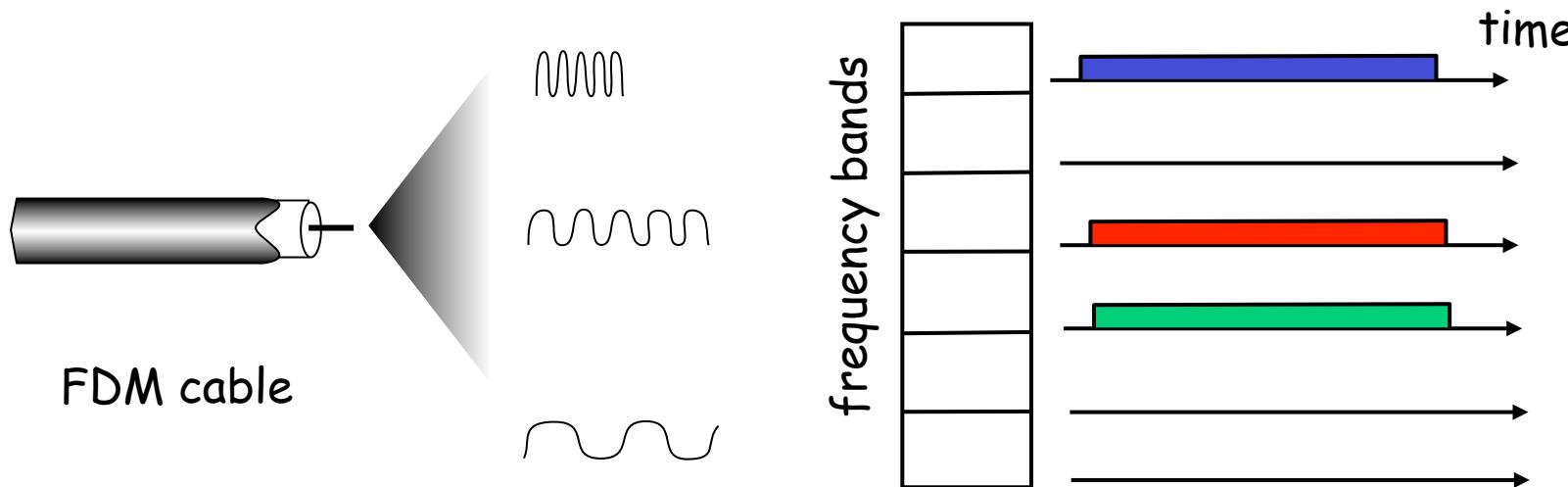
- access to channel in "rounds"
- each station gets fixed length slot (length = pkt trans time) in each round
- unused slots go idle
- example: 6-station LAN, 1,3,4 have pkt, slots 2,5,6 idle



Channel Partitioning MAC protocols: FDMA

FDMA: frequency division multiple access

- channel spectrum divided into frequency bands
- each station assigned fixed frequency band
- unused transmission time in frequency bands go idle
- example: 6-station LAN, 1,3,4 have pkt, frequency bands 2,5,6 idle



TDMA/FDMA Vs. Ideal Multiple Access Protocol

Broadcast channel of rate R bps

1. when one node wants to transmit, it can send at rate R. → NOT MET BY TDMA/FDMA
2. when M nodes want to transmit, each can send at average rate R/M → MET BY TDMA/FDMA
3. fully decentralized:
 - no special node to coordinate transmissions
 - no synchronization of clocks, slots
4. simple

Random Access Protocols

- When node has packet to send
 - transmit at full channel data rate R.
 - no *a priori* coordination among nodes
- two or more transmitting nodes → “collision”,
- random access MAC protocol specifies:
 - how to detect collisions
 - how to recover from collisions (e.g., via delayed retransmissions)
- Examples of random access MAC protocols:
 - slotted ALOHA
 - ALOHA
 - CSMA, CSMA/CD, CSMA/CA

Slotted ALOHA

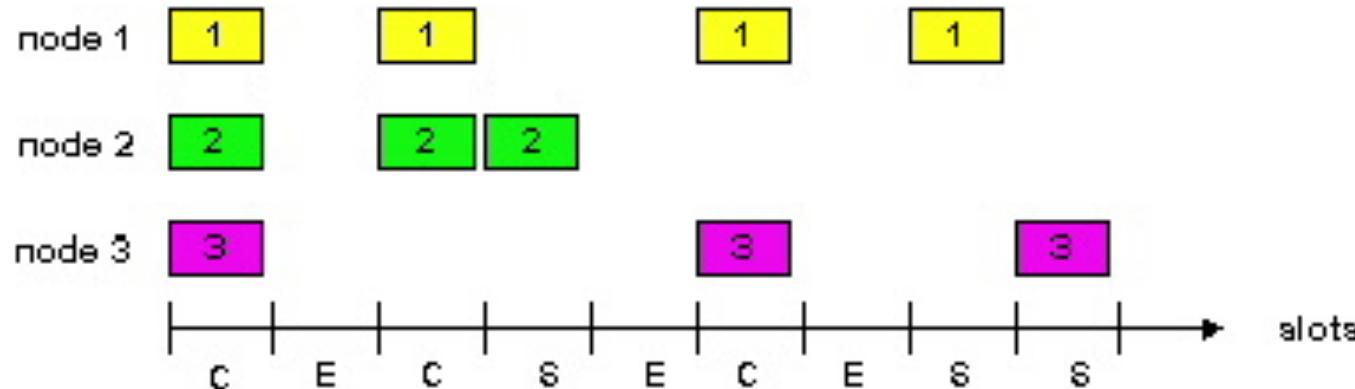
Assumptions:

- all frames same size
- time divided into equal size slots (time to transmit 1 frame)
- nodes start to transmit only slot beginning
- nodes are synchronized
- if 2 or more nodes transmit in slot, all nodes detect collision

Operation:

- when node obtains fresh frame, transmits in next slot
 - if no collision:* node can send new frame in next slot
 - if collision:* node retransmits frame in each subsequent slot with prob. p until success

Slotted ALOHA



Pros

- single active node can continuously transmit at full rate of channel
- highly decentralized: only slots in nodes need to be in sync
- simple

Cons

- collisions, wasting slots
- idle slots
- nodes may be able to detect collision in less than time to transmit packet
- clock synchronization

Slotted Aloha efficiency

Efficiency : long-run fraction of successful slots (many nodes, all with many frames to send)

- suppose: N nodes with many frames to send, each transmits in slot with probability p
- prob that given node has success in a slot = $p(1-p)^{N-1}$
- prob that *any* node has a success = $Np(1-p)^{N-1}$

- max efficiency: find p^* that maximizes $Np(1-p)^{N-1}$
- for many nodes, take limit of $Np^*(1-p^*)^{N-1}$ as N goes to infinity, gives:

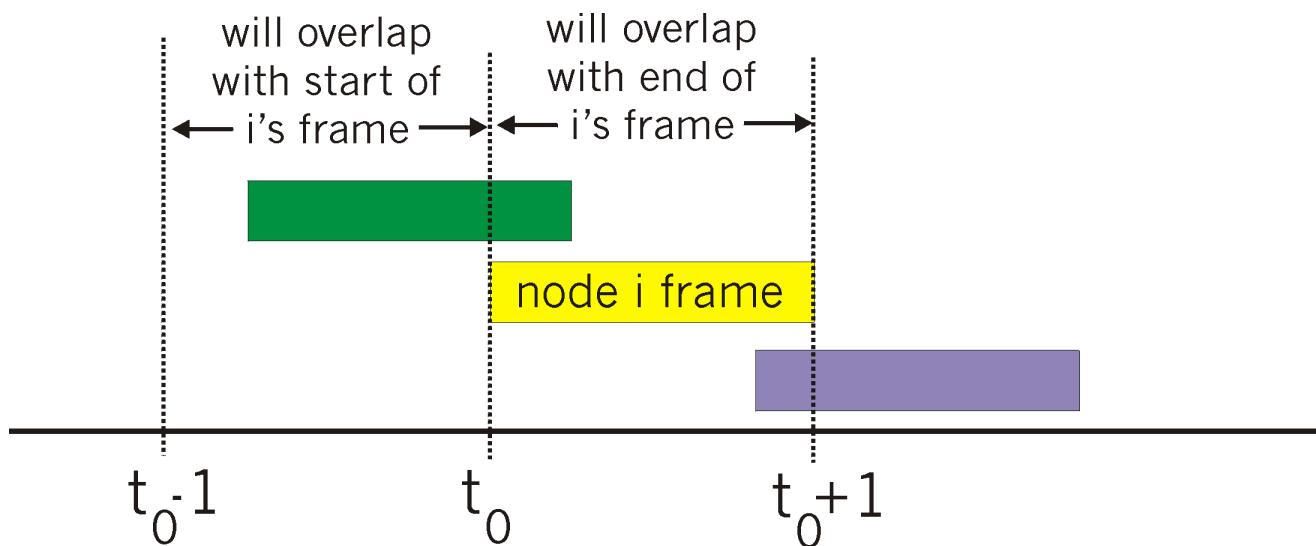
$$\text{Max efficiency} = 1/e = .37$$

At best: channel used for useful transmissions 37% of time!

!

Pure (unslotted) ALOHA

- unslotted Aloha: simpler, no synchronization
- when frame first arrives
 - transmit immediately
- collision probability increases:
 - frame sent at t_0 collides with other frames sent in $[t_0-1, t_0+1]$



Pure Aloha efficiency

$P(\text{success by given node}) = P(\text{node transmits}) \cdot$

$P(\text{no other node transmits in } [p_0-1, p_0]) \cdot$

$P(\text{no other node transmits in } [p_0-1, p_0])$

$$= p \cdot (1-p)^{N-1} \cdot (1-p)^{N-1}$$

$$= p \cdot (1-p)^{2(N-1)}$$

... choosing optimum p and then letting $n \rightarrow \infty$...

$$= 1/(2e) = .18$$

even worse than slotted Aloha!