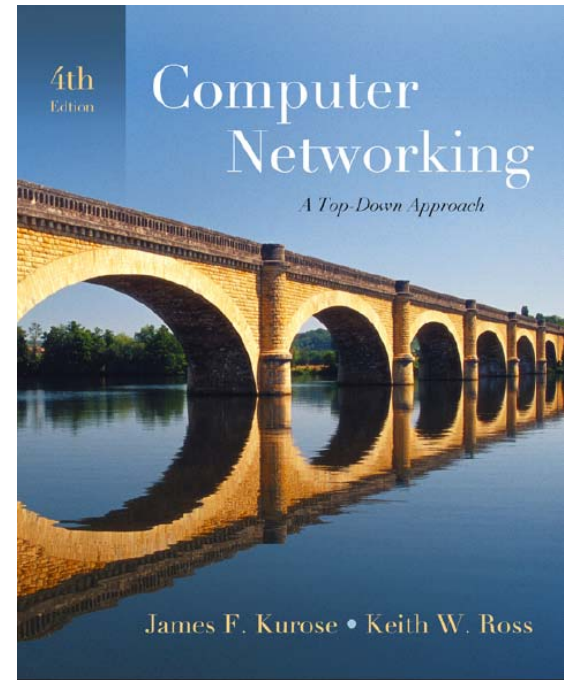


# 第一章 計算機網路與網際網路 Chapter 1 Introduction



*Computer Networking:  
A Top Down Approach*

4<sup>th</sup> edition.

Jim Kurose, Keith Ross  
Addison-Wesley, July  
2007.

# Chapter 1: Introduction

## Our goal:

- ❑ get “feel” and terminology
- ❑ more depth, detail *later* in course
- ❑ approach:
  - ❖ use Internet as example

## Overview:

- ❑ 什麼是網際網路 ( Internet ) ?
- ❑ 什麼是協定 ( Protocol ) ?
- ❑ 網路的邊際 ( network edge )
- ❑ 網路的核心 ( network core )
- ❑ 網路效能 ( performance )
  - ❖ 封包遺失 ( loss )
  - ❖ 延遲 ( delay )
  - ❖ 產出量 ( throughput )
- ❑ 安全 ( security )
- ❑ 協定分層 ( protocol layers ) , service models
- ❑ 網路的歷史 ( history )

# Chapter 1: roadmap

1.1 What *is* the Internet? 什麼是網際網路？

1.2 Network edge

- end systems, access networks, links

1.3 Network core

- circuit switching, packet switching, network structure

1.4 Delay, loss and throughput in packet-switched networks

1.5 Protocol layers, service models



1.6 Networks under attack: security

1.7 History

# 什麼是網際網路？從基本元素來看

-  PC
  -  server
  -  wireless laptop
  -  cellular handheld
- millions of connected computing devices:  
主機 (終端系統)  
*hosts = end systems*
- ❖ 執行應用程式  
running *network apps*

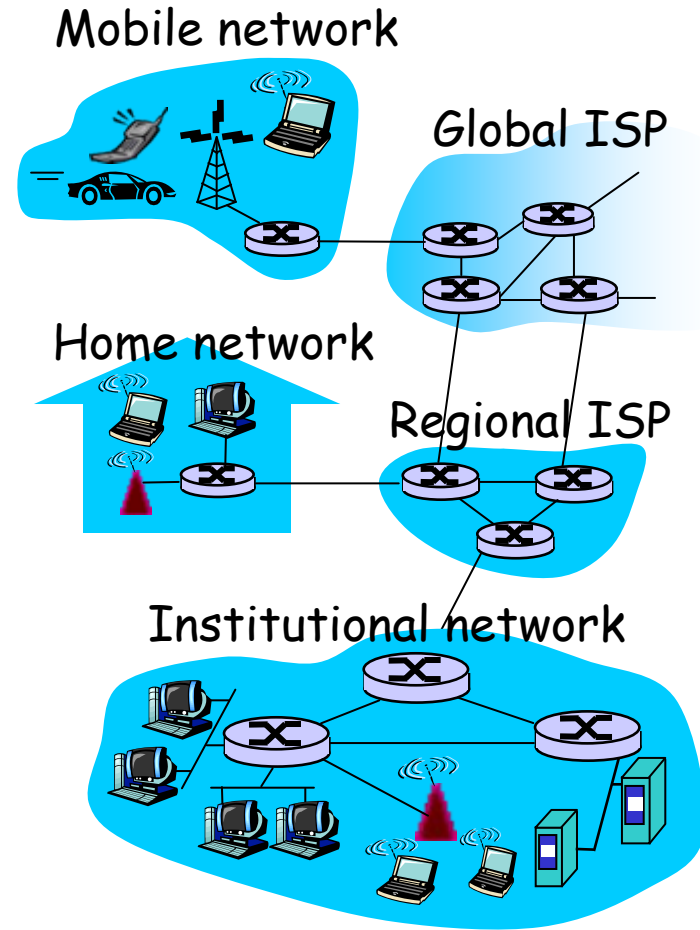
## □ *communication links*

-  access points
-  wired links

- ❖ 光纖、纜線、衛星...
- ❖ 頻寬 (bandwidth) ?  
transmission rate = *bandwidth*

-  router

- 路由器 (Routers) :  
forward packets  
(chunks of data)



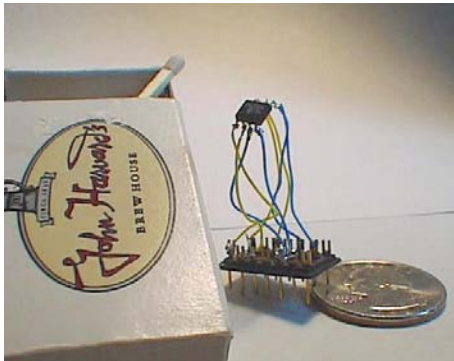
# “Cool” internet appliances



IP picture frame  
<http://www.ceiva.com/>



Web-enabled toaster +  
weather forecaster



World's smallest web server  
<http://www-ccs.cs.umass.edu/~shri/iPic.html>



Internet phones

# What's a protocol ( 通訊協定 ) ?

## human protocols:

### 人類的通訊協定

- ❑ “what’ s the time?”
- ❑ “I have a question”
- ❑ introductions

... specific msgs sent

... specific actions taken  
when msgs received, or  
other events

## network protocols:

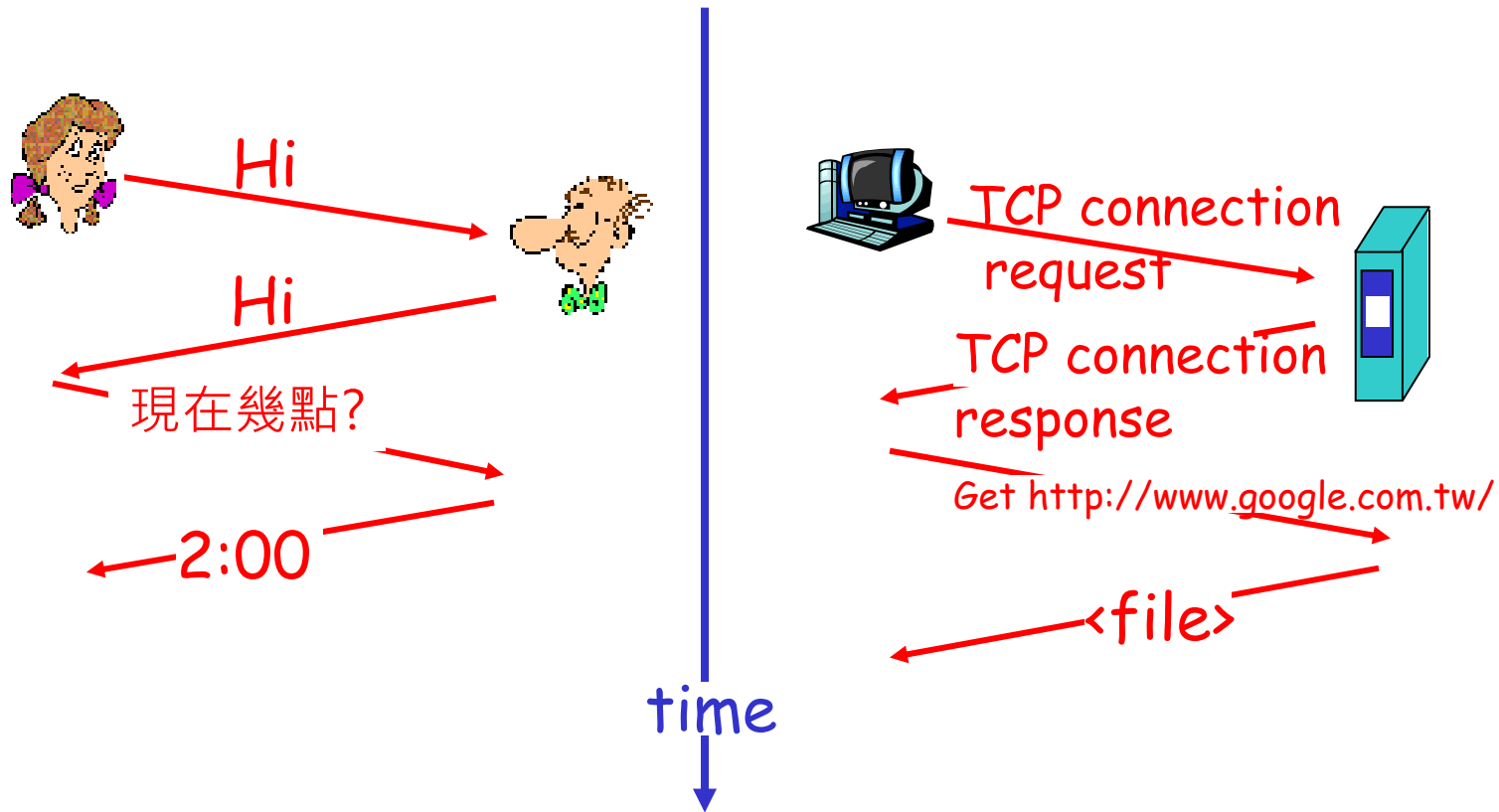
### 網路通訊協定

- ❑ machines rather than humans (no ambiguous)
- ❑ all communication activity in Internet governed by protocols

*protocols define format,  
order of msgs sent and  
received among network  
entities, and actions taken  
on msg transmission,  
receipt*

# What's a protocol ( 通訊協定 ) ?

## 人類協定與電腦網路協定的異同



Q: Other human protocols?  
有沒有其它的人類協定？

# Chapter 1: roadmap

1.1 What *is* the Internet?

1.2 Network edge 網路邊際

- end systems, access networks, links

- 終端系統、接取網路、鏈結

1.3 Network core

- circuit switching, packet switching, network structure

1.4 Delay, loss and throughput in packet-switched networks

1.5 Protocol layers, service models

1.6 Networks under attack: security

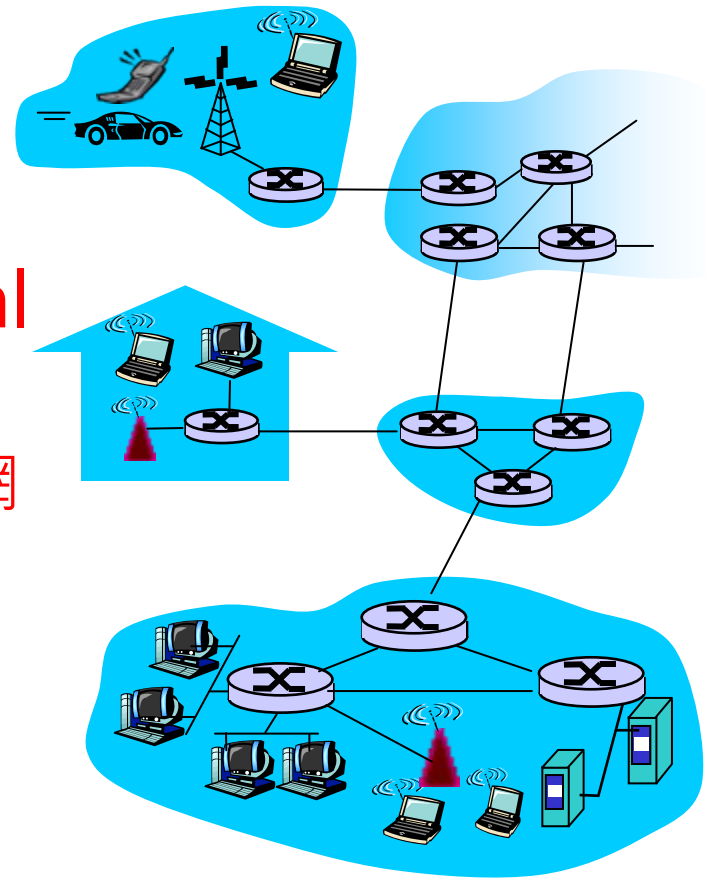
1.7 History



# A closer look at network structure:

## 細看網路架構

- **network edge:** applications and hosts (應用程式及主機)
- **access networks, physical media:** wired, wireless communication links (網路連線)
- **network core:** 網路核心
  - ❖ interconnected routers
  - ❖ network of networks



# Network edge: reliable data transfer service

## 可信賴的傳輸服務

- Goal: data transfer  
between end systems
- ❑ *handshaking*: setup (prepare for) data transfer ahead of time
    - ❖ Hello, hello back human protocol
    - ❖ *set up "state"* in two communicating hosts
  - ❑ **TCP** - Transmission Control Protocol
    - ❖ Internet's reliable data transfer service

- 補充 :
- TCP service [RFC 793]
- ❑ *reliable, in-order* byte-stream data transfer
    - ❖ loss: acknowledgements and retransmissions
  - ❑ *flow control*:
    - ❖ sender won't overwhelm receiver
  - ❑ *congestion control*:
    - ❖ senders "slow down sending rate" when network congested

# Network edge: best effort (unreliable) data transfer service 盡力而為的服務模式

Goal: data transfer  
between end systems

- ❖ same as before!

- ❑ **UDP** - User Datagram Protocol [RFC 768]:

- ❖ connectionless

- ❖ **unreliable** data transfer

- ❖ **no flow control**

- ❖ **no congestion control**

App's using TCP:

- ❑ HTTP (Web), FTP (file transfer), Telnet (remote login), SMTP (email)

App's using UDP:

- ❑ streaming media, teleconferencing, DNS, Internet telephony

# Access networks and physical media

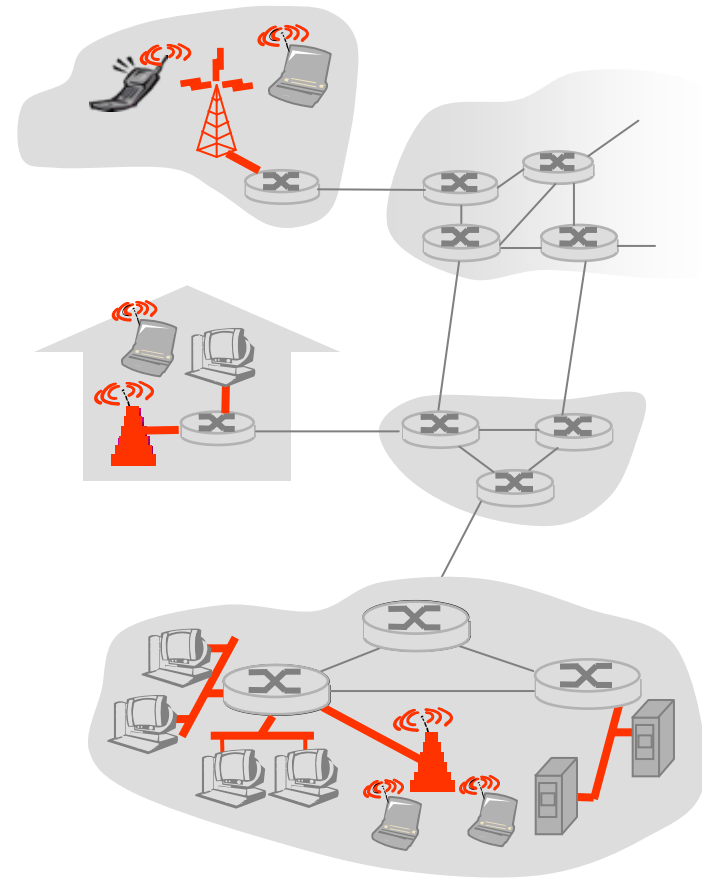
## 連接網路與實體媒介

*Q: How to connect end systems to edge router?*

- ❑ residential access nets
- ❑ institutional access networks (school, company)
- ❑ mobile access networks

*Keep in mind:*

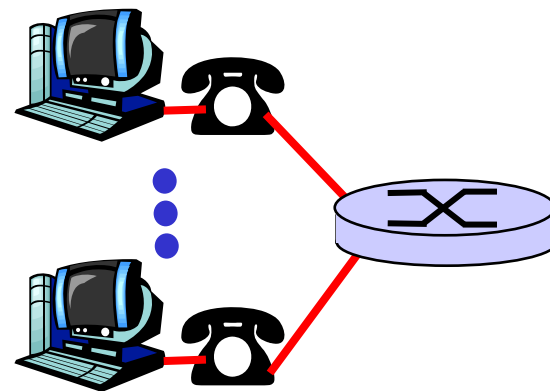
- ❑ bandwidth (bits per second) of access network? 頻寬有多大?
- ❑ shared or dedicated? 共用或專用?



# Residential access: point to point access

## ❑ Dialup via modem 透過數據機撥接

- ❖ up to 56Kbps direct access to router (often less)
- ❖ Can't surf and phone at same time: can't be "always on"



## ❑ DSL: digital subscriber line 數位用戶專線

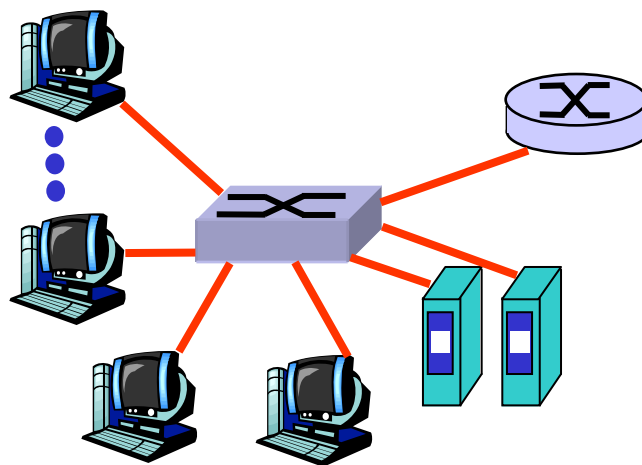
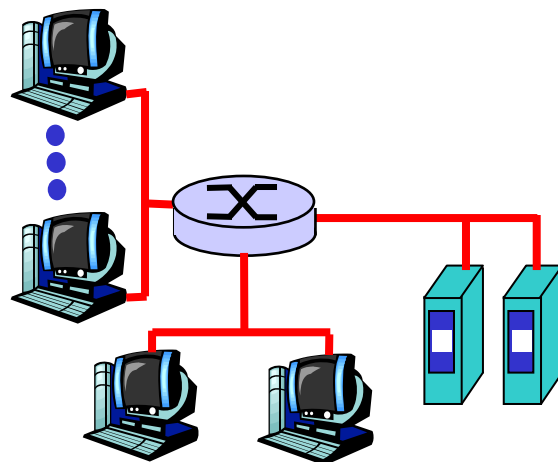
- ❖ deployment: telephone company (typically)
- ❖ up to 1 Mbps upstream (today typically < 256 kbps)
- ❖ up to 8 Mbps downstream (today typically < 1 Mbps)
- ❖ dedicated physical line to telephone central office
- ❖ ADSL (Asymmetric Digital Subscriber Line)

# Residential access: cable modems

- ❑ HFC: hybrid fiber-coaxial cable 混合光纖同軸電纜
  - ❖ asymmetric: up to 30Mbps downstream, 2 Mbps upstream
- ❑ network of cable and fiber attaches homes to ISP router
  - ❖ homes share access to router 共享
- ❑ deployment: available via cable TV companies

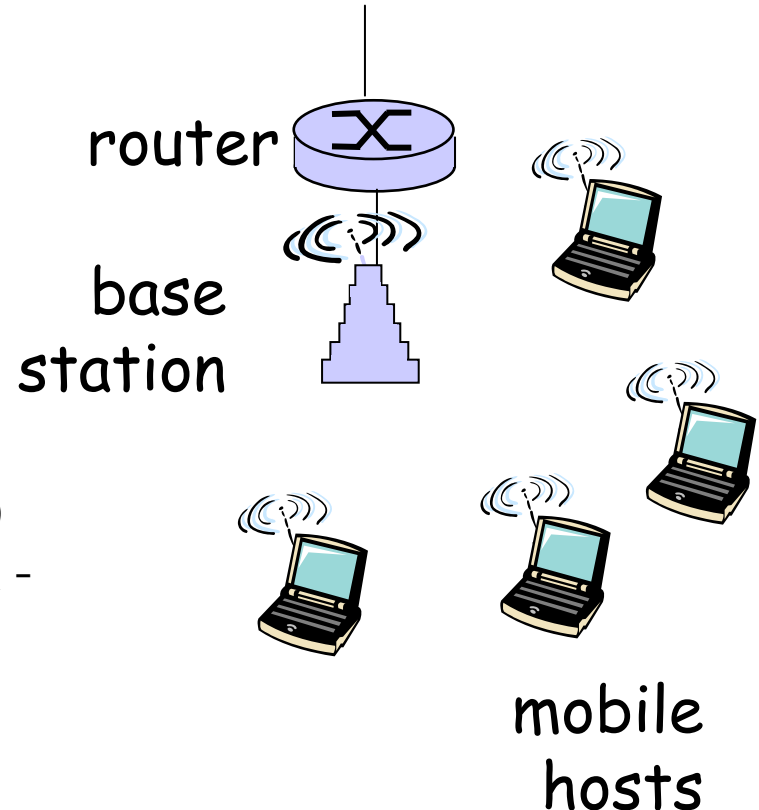
# Company access: local area networks

- ❑ company/univ **local area network** (LAN 區域網路) connects end system to edge router
- ❑ **Ethernet**: 乙太網路
  - ❖ 10 Mbps, 100Mbps, 1Gbps, 10Gbps Ethernet
  - ❖ modern configuration: end systems connect into *Ethernet switch*
- ❑ LANs: chapter 5



# Wireless access networks 無線網路

- **shared wireless** access network connects end system to router
  - ❖ via base station 基地台 aka “access point ( AP ) ”
- **wireless LANs:**
  - ❖ 802.11b/g (WiFi): 11 or 54 Mbps
- **wide-area wireless access**
  - ❖ provided by telecom operator
  - ❖ ~1Mbps over cellular system (EVDO - Evolution-Data Optimized, HSDPA - High-Speed Downlink Packet)
  - ❖ next up (?): 5-10Mbps, WiMAX IEEE 802.16(10' s Mbps) over wide area

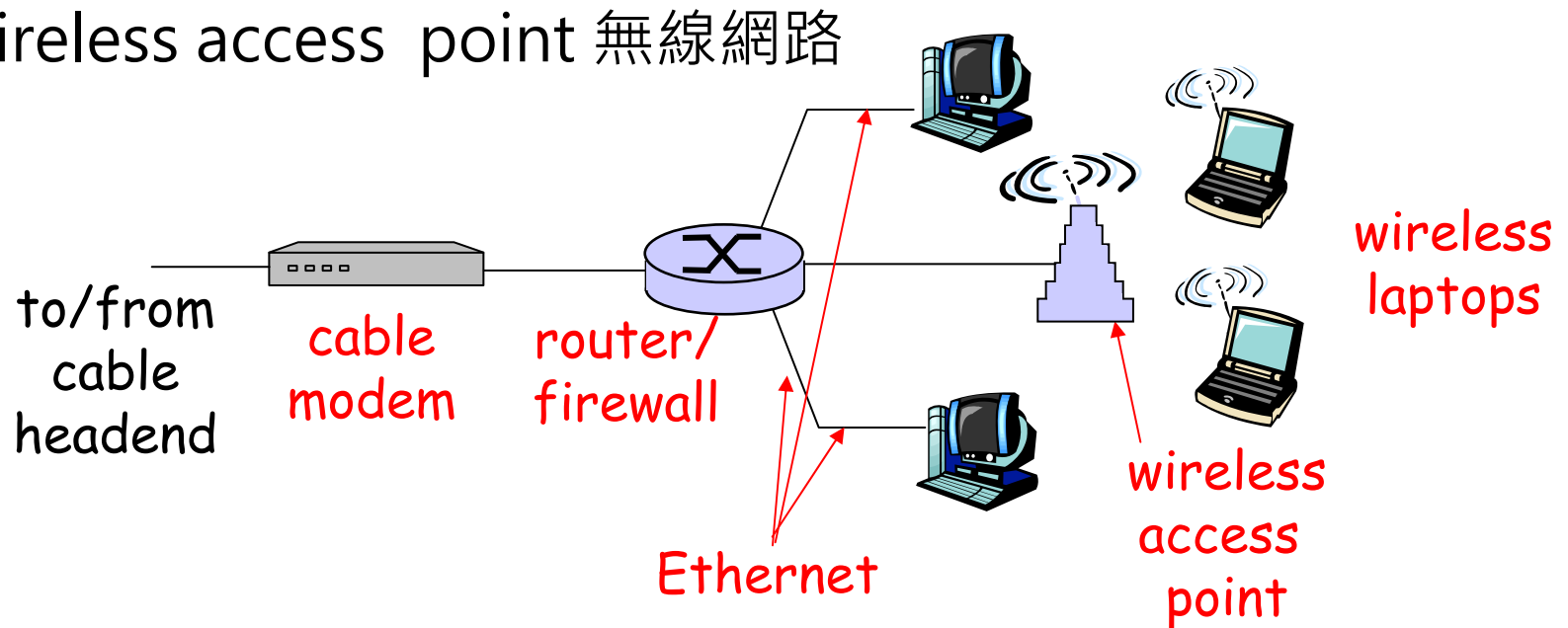




# Home networks 家庭網路

## Typical home network components:

- ❑ DSL or cable modem 數據機
- ❑ router/firewall/NAT 路由器、防火牆、IP 分享器
- ❑ Ethernet 乙太網路
- ❑ wireless access point 無線網路



# Physical Media 實體媒介

- ❑ **bit**: propagates between transmitter/rcvr pairs
- ❑ **physical link**: what lies between transmitter & receiver
- ❑ **guided media**: 導引式媒介
  - ❖ signals propagate in solid media: copper, fiber, coax
- ❑ **unguided media**: 非導引式媒介
  - ❖ signals propagate freely, e.g., radio

## Twisted Pair (TP) 雙絞線

- ❑ two insulated copper wires
  - ❖ Category 3: traditional phone wires, 10 Mbps Ethernet
  - ❖ Category 5: 100Mbps Ethernet



# Twisted Pair (TP) 雙絞線

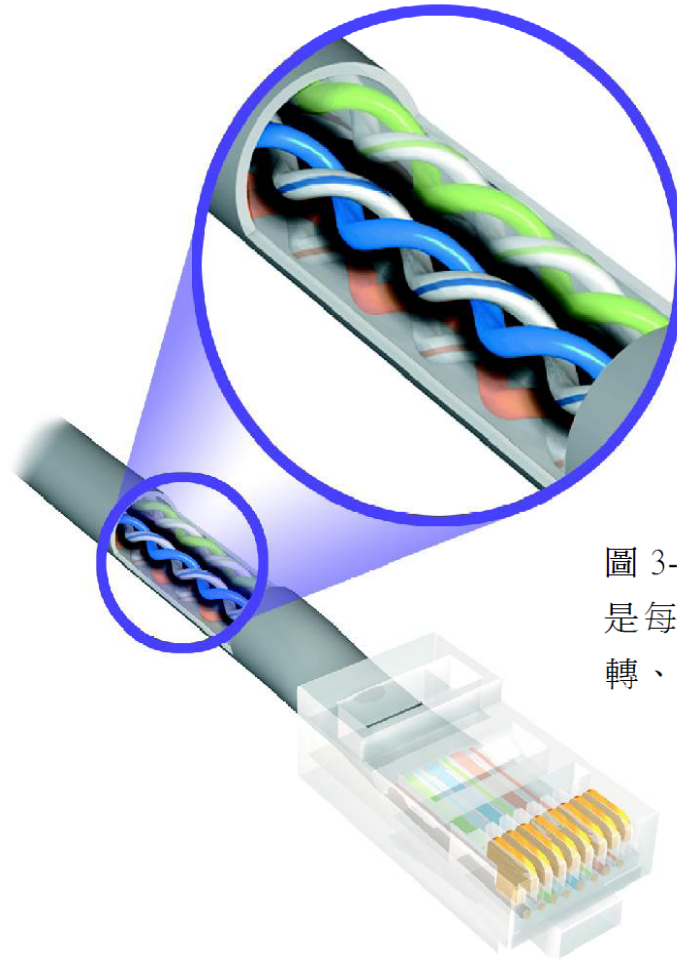
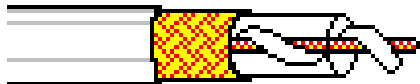


圖 3-03 雙絞線的特色  
是每兩條芯線相互旋  
轉、纏繞在一起

# Physical Media: coax, fiber 同軸電纜、光纖

## Coaxial cable:

- ❑ two concentric copper conductors
- ❑ bidirectional
- ❑ baseband:
  - ❖ single channel on cable
  - ❖ legacy Ethernet
- ❑ broadband:
  - ❖ multiple channels on cable
  - ❖ HFC



## Fiber optic cable:

- ❑ glass fiber carrying light pulses, each pulse a bit
- ❑ high-speed operation:
  - ❖ high-speed point-to-point transmission (e.g., 10's-100's Gps)
- ❑ low error rate: repeaters spaced far apart ; immune to electrical noise



# Physical media: radio 無線電

- ❑ signal carried in electromagnetic spectrum
- ❑ no physical “wire”
- ❑ bidirectional
- ❑ propagation environment effects:
  - ❖ reflection
  - ❖ obstruction by objects
  - ❖ interference

## Radio link types:

- ❑ **terrestrial microwave**
  - ❖ e.g. up to 45 Mbps channels
- ❑ **LAN** (e.g., Wifi)
  - ❖ 11Mbps, 54 Mbps
- ❑ **wide-area** (e.g., cellular)
  - ❖ 3G cellular: ~ 1 Mbps
- ❑ **satellite**
  - ❖ Kbps to 45Mbps channel (or multiple smaller channels)
  - ❖ 270 msec end-end delay
  - ❖ geosynchronous versus low altitude

# Chapter 1: roadmap

1.1 What *is* the Internet?

1.2 Network edge

- end systems, access networks, links

1.3 Network core

- circuit switching, packet switching, network structure

1.4 Delay, loss and throughput in packet-switched networks

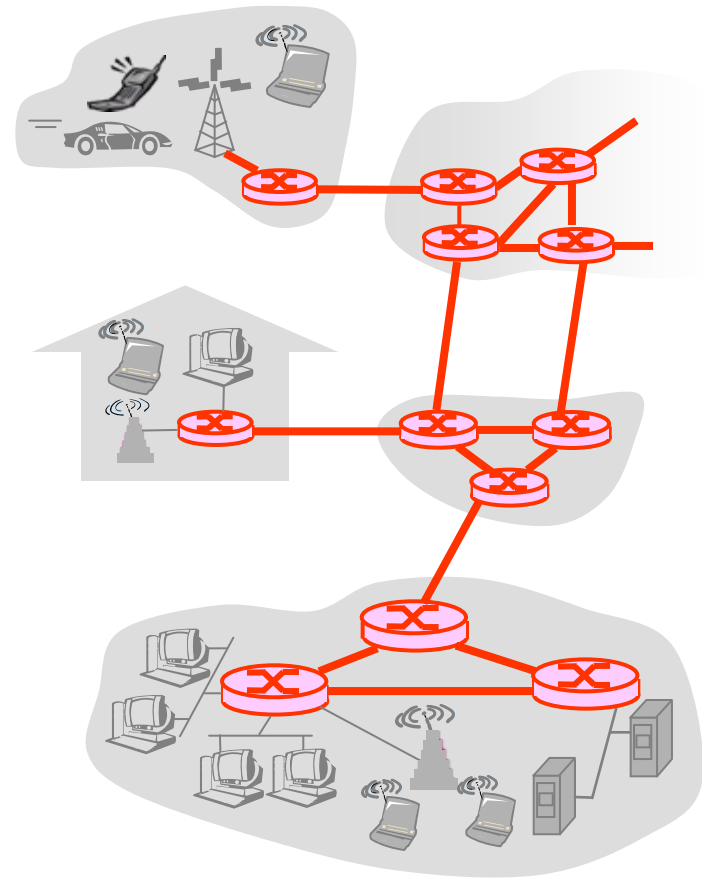
1.5 Protocol layers, service models

1.6 Networks under attack: security

1.7 History

# The Network Core 網路核心

- ❑ mesh of interconnected routers
- ❑ the fundamental question: how is data transferred through net?
  - ❖ **circuit switching**: 電路交換 dedicated circuit per call: telephone net
  - ❖ **packet-switching**: 封包交換 data sent thru net in discrete "chunks"



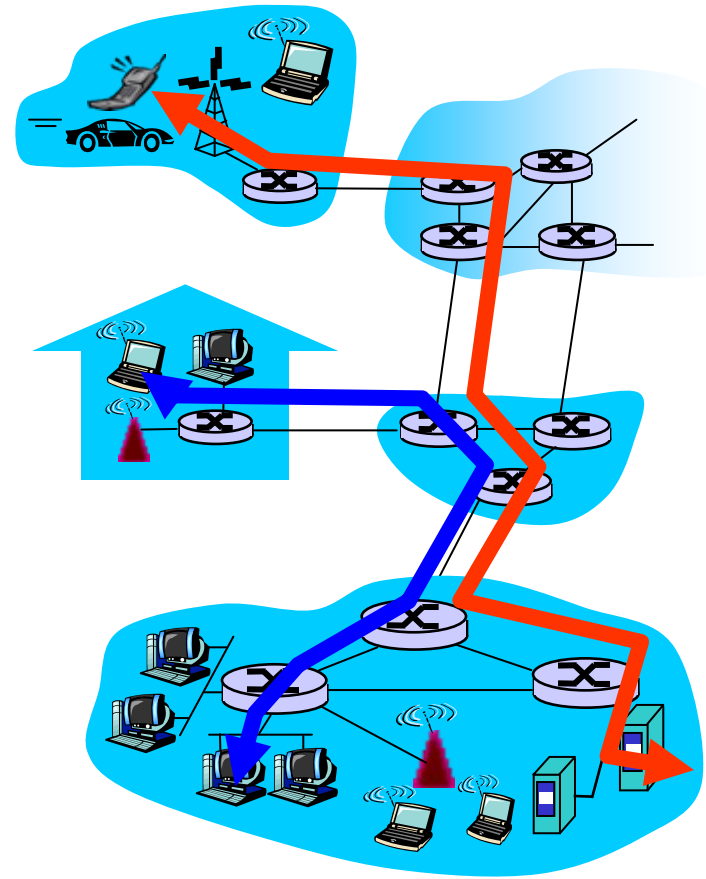
# Network Core: Circuit Switching

## 網路核心：電路交換

End-end resources reserved for "call"

資源保留

- ❑ link bandwidth, switch capacity
- ❑ dedicated resources: no sharing
- ❑ circuit-like (guaranteed) performance
- ❑ call setup required





# Network Core: Circuit Switching

## 網路核心：電路交換

network resources

(e.g., bandwidth)

divided into "pieces"

- pieces allocated to calls
- resource piece *idle* if not used by owning call (*no sharing*)

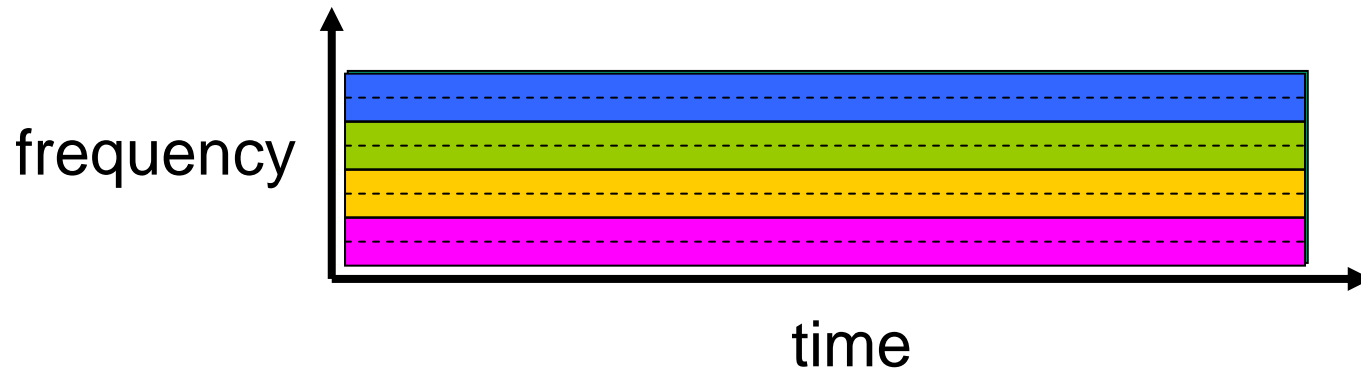
- dividing link bandwidth into "pieces"
  - ❖ frequency division
  - ❖ time division

# Circuit Switching: FDM and TDM

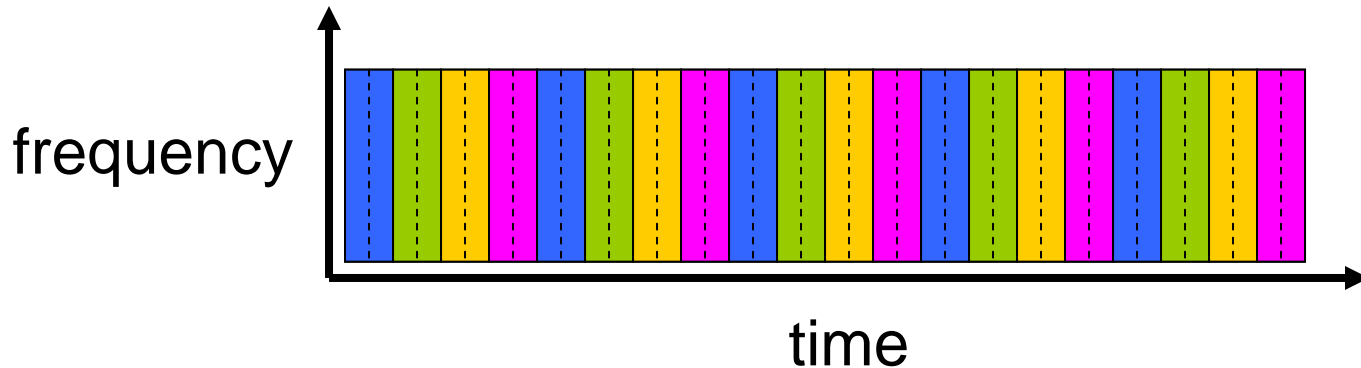
FDM : circuit=frequency

Example:

4 users



TDM : circuit=time slot



# Numerical example

- How long does it take to send a file of 640,000 bits from host A to host B over a circuit-switched network?
  - ❖ All links are 1.536 Mbps
  - ❖ Each link uses TDM with 24 slots/sec
  - ❖ 500 msec to establish end-to-end circuit

Let's work it out!

ANS : 10.5 sec

# Network Core: Packet Switching

## 網路核心：封包交換

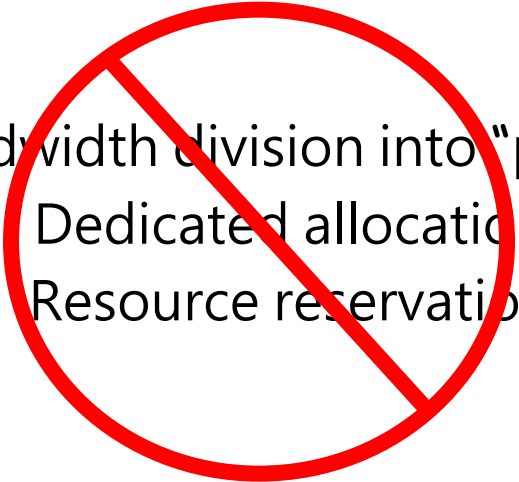
each end-end data stream  
divided into *packets*

- ❑ user A, B packets *share* network resources
- ❑ each packet uses full link bandwidth
- ❑ resources used *as needed*

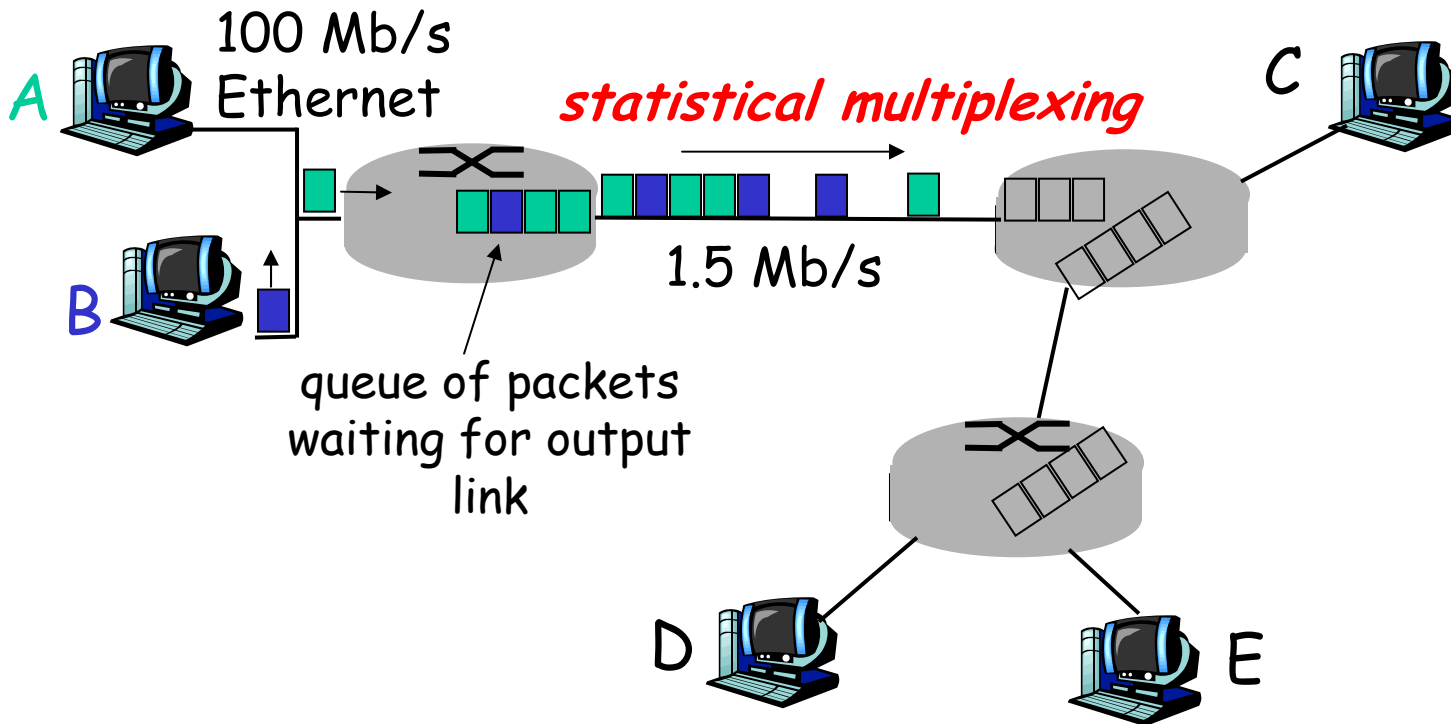
resource contention:

- ❑ aggregate resource demand can exceed amount available
- ❑ congestion: packets queue, wait for link use
- ❑ store and forward: packets move one hop at a time
  - ❖ Node receives complete packet before forwarding

Bandwidth division into "pieces"  
Dedicated allocation  
Resource reservation



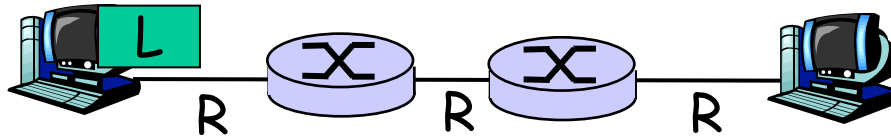
# Packet Switching: Statistical Multiplexing



Sequence of A & B packets does not have fixed pattern, bandwidth shared on demand → *statistical multiplexing*.

TDM: each host gets same slot in revolving TDM frame.

# Packet-switching: store-and-forward



- takes  $L/R$  seconds to transmit (push out) packet of  $L$  bits on to link at  $R$  bps 傳輸時間
- *store and forward*: entire packet must arrive at router before it can be transmitted on next link 先收完再送
- delay =  $3L/R$  (assuming zero propagation delay)

## Example:

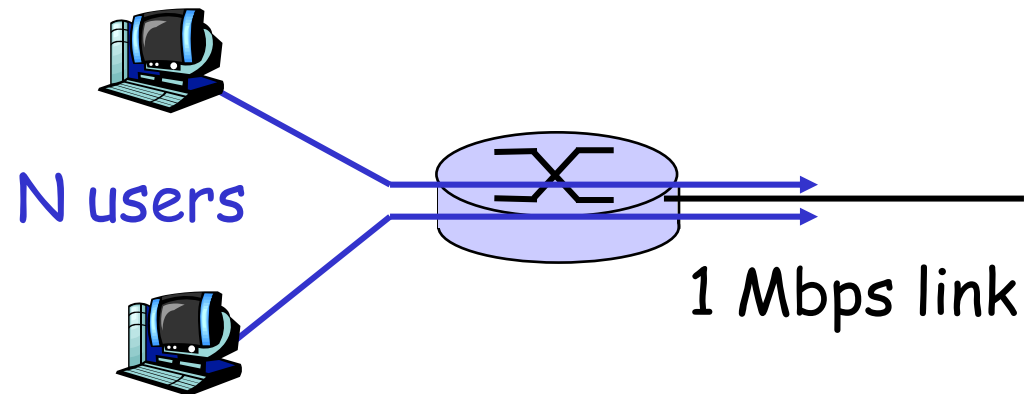
- $L = 7.5$  Mbits
- $R = 1.5$  Mbps
- transmission delay = 15 sec

} more on delay shortly ...

# Packet switching versus circuit switching

*Packet switching allows more users to use network!*  
*封包交換使更多使用者使用網路*

- ❑ 1 Mb/s link
- ❑ each user:
  - ❖ 100 kb/s when "active"
  - ❖ active 10% of time
- ❑ *circuit-switching:*
  - ❖ 10 users ( $1\text{M}/100\text{k}=10$ )
- ❑ *packet switching:*
  - ❖ with 35 users, probability  $> 10$  active at same time is about .0004



Q: how did we get value 0.0004?

# Packet switching versus circuit switching

The screenshot shows the KDevelop IDE with a C++ file named `cal.cpp` open. The code defines a factorial function, a combination function, and a main function that calculates a sum. The terminal output shows the program's execution result.

```
file:///home/synn/cal.cpp - KDevelop
檔案(E) 編輯(E) 檢視(V) 專案(P) 建立(I) 除錯(D) 文稿(C) 書籤(B) 視窗(W) 工具(T) 設定(S) 說明(H)

Simulation_new.cpp cal.cpp
#include <iostream.h>
#include <math.h>

double fact(int n)
{
    if(n==0)
        return 1.0;
    else
        return (double)n*fact(n-1);
}

double combination(int n, int i)
{
    return fact(n)/(fact(i)*fact(n-i));
}

int main()
{
    double sum=0.0;
    for(int i=0;i<=10;i++)
        sum=sum+combination(35,i)*pow(0.1,i)*pow(0.9,35-i);
    cout<<1.0-sum<<endl;
    return 0;
}

[synn@alpha25 ~]$ ./cal
0.000424298
[synn@alpha25 ~]$ ~
```

行: 14欄: 24 插入 一般

收件匣 - synnwang@gmail.com  
file:///home/synn/cal.cpp - K

09:46  
2009-09-21



# Packet switching versus circuit switching

Is packet switching a “slam dunk winner?”

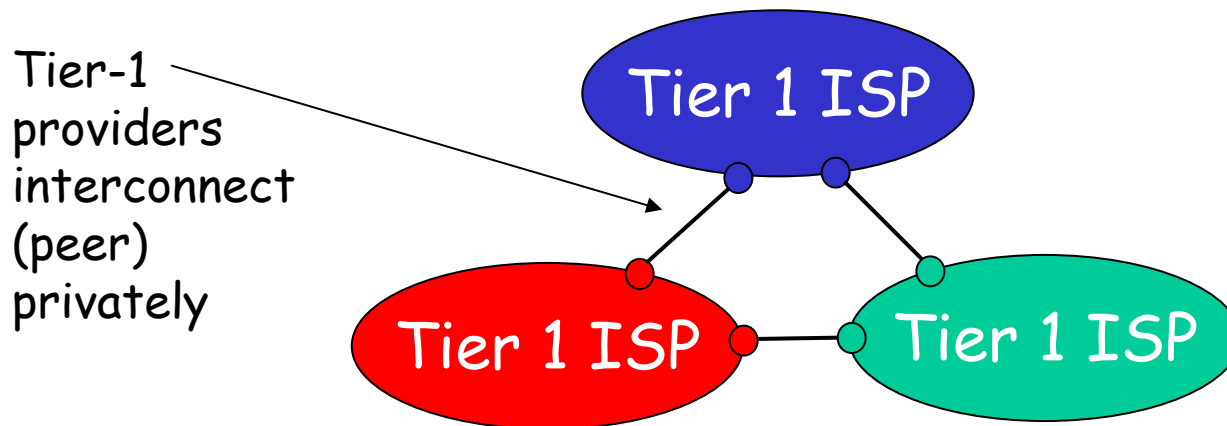
- ❑ great for bursty data
  - ❖ resource sharing
  - ❖ simpler, no call setup
- ❑ **excessive congestion**: packet delay and loss
  - ❖ protocols needed for reliable data transfer, congestion control
- ❑ **Q: How to provide circuit-like behavior?**
  - ❖ bandwidth guarantees needed for audio/video apps
  - ❖ still an unsolved problem (chapter 7)

Q: human analogies of reserved resources (circuit switching) versus on-demand allocation (packet-switching)?

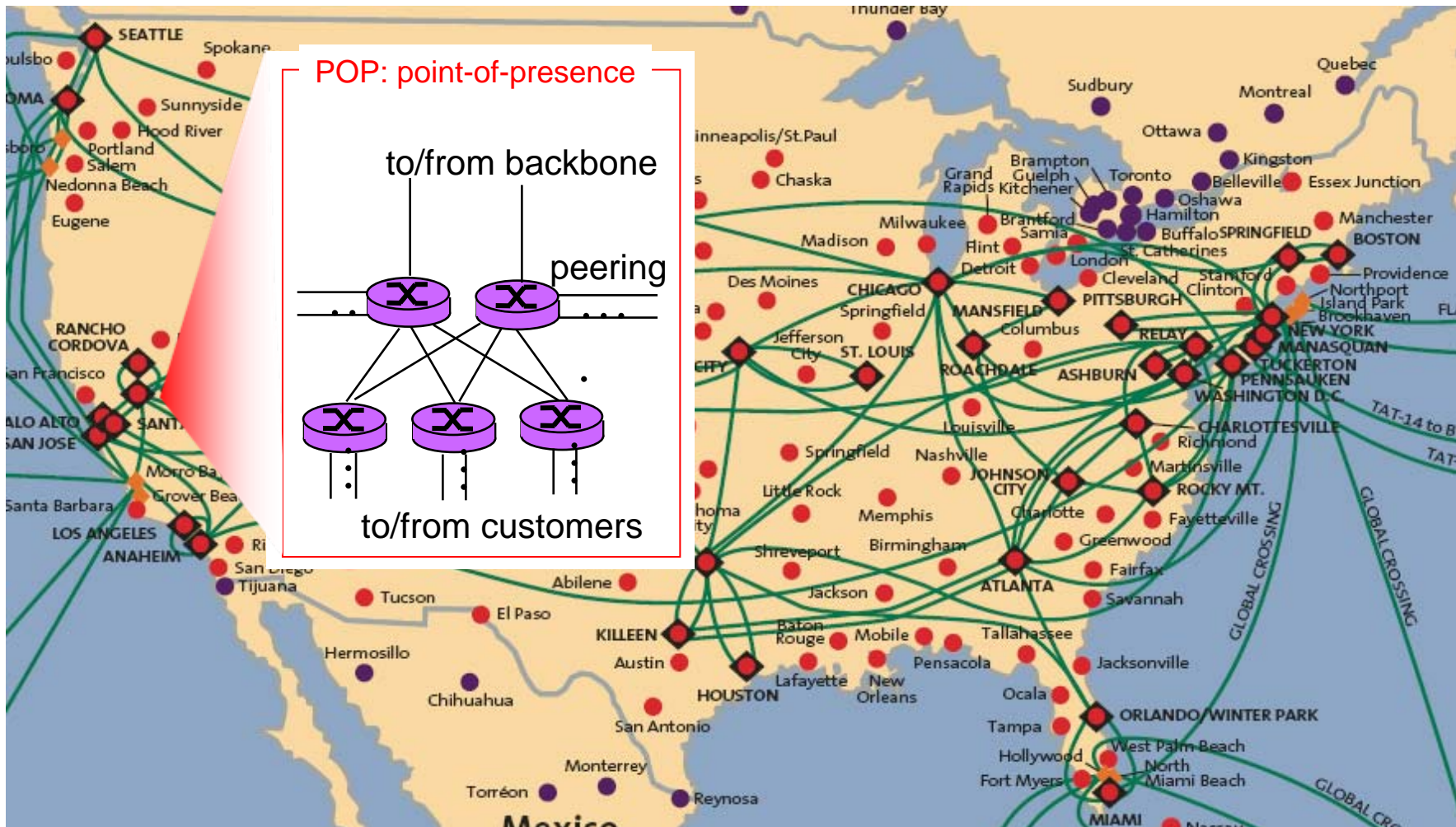
# Internet structure: network of networks

## 網際網路架構：網路的網路...

- roughly hierarchical 階層式架構
- **at center:** “tier-1 第一層” ISPs (e.g., Verizon, Sprint, AT&T, Cable and Wireless), national/international coverage
  - ❖ treat each other as equals

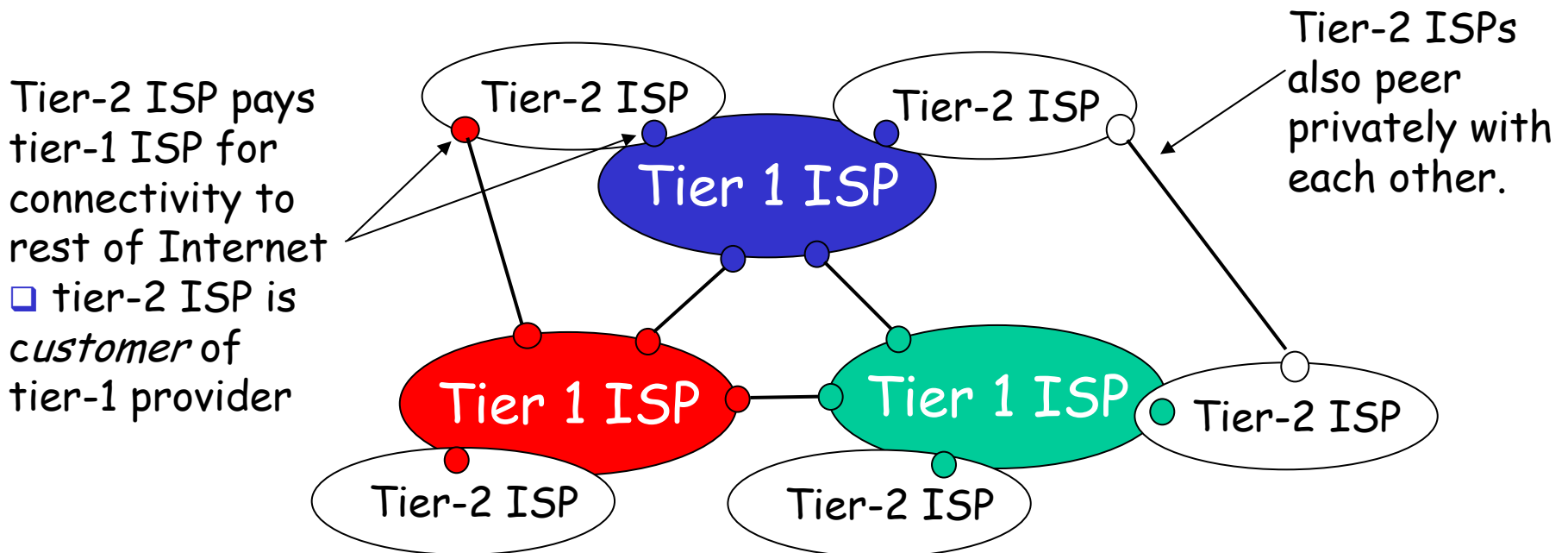


# Tier-1 ISP: e.g., Sprint



# Internet structure: network of networks

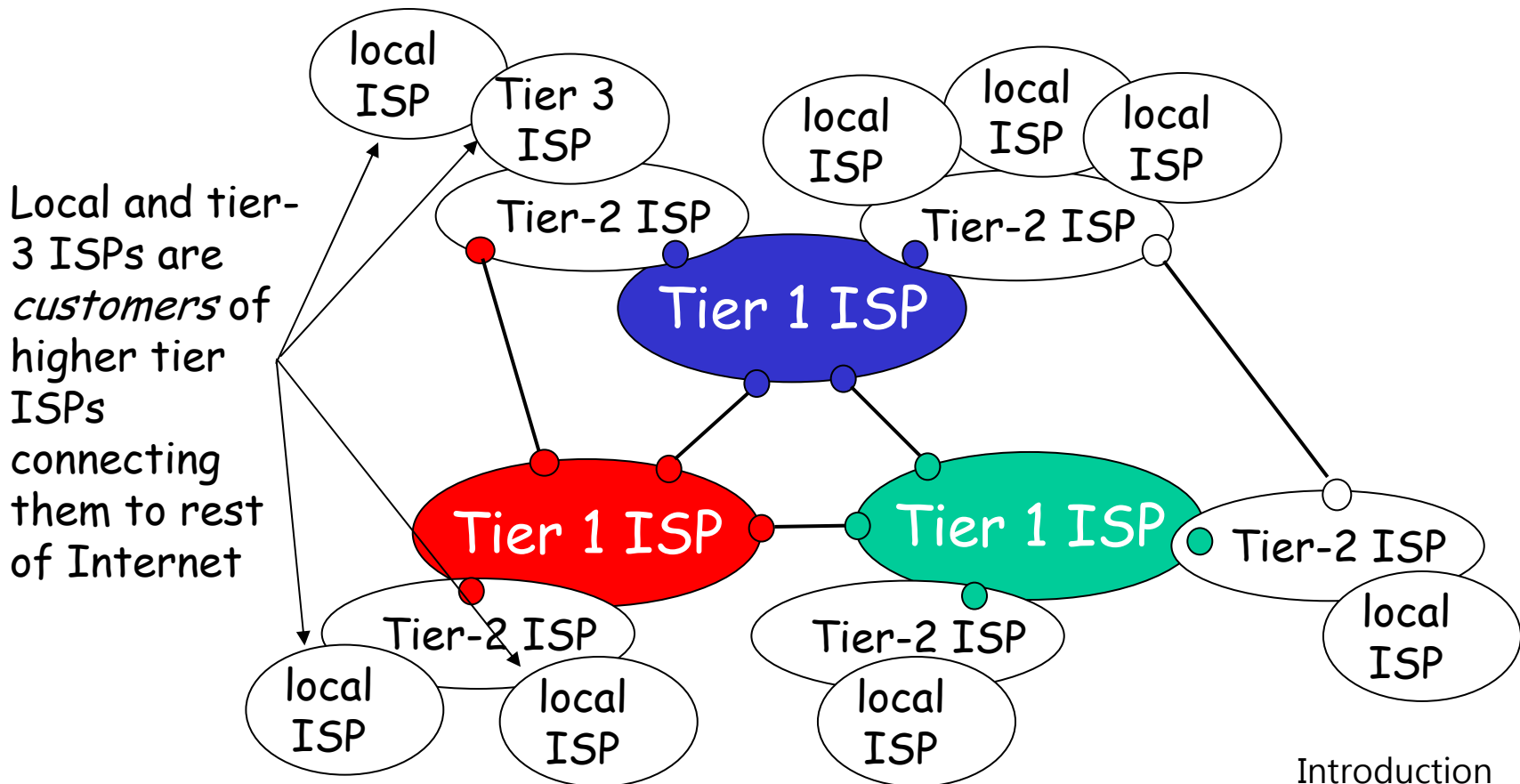
- “Tier-2 第二層” ISPs: smaller (often regional) ISPs
  - ❖ Connect to one or more tier-1 ISPs, possibly other tier-2 ISPs



# Internet structure: network of networks

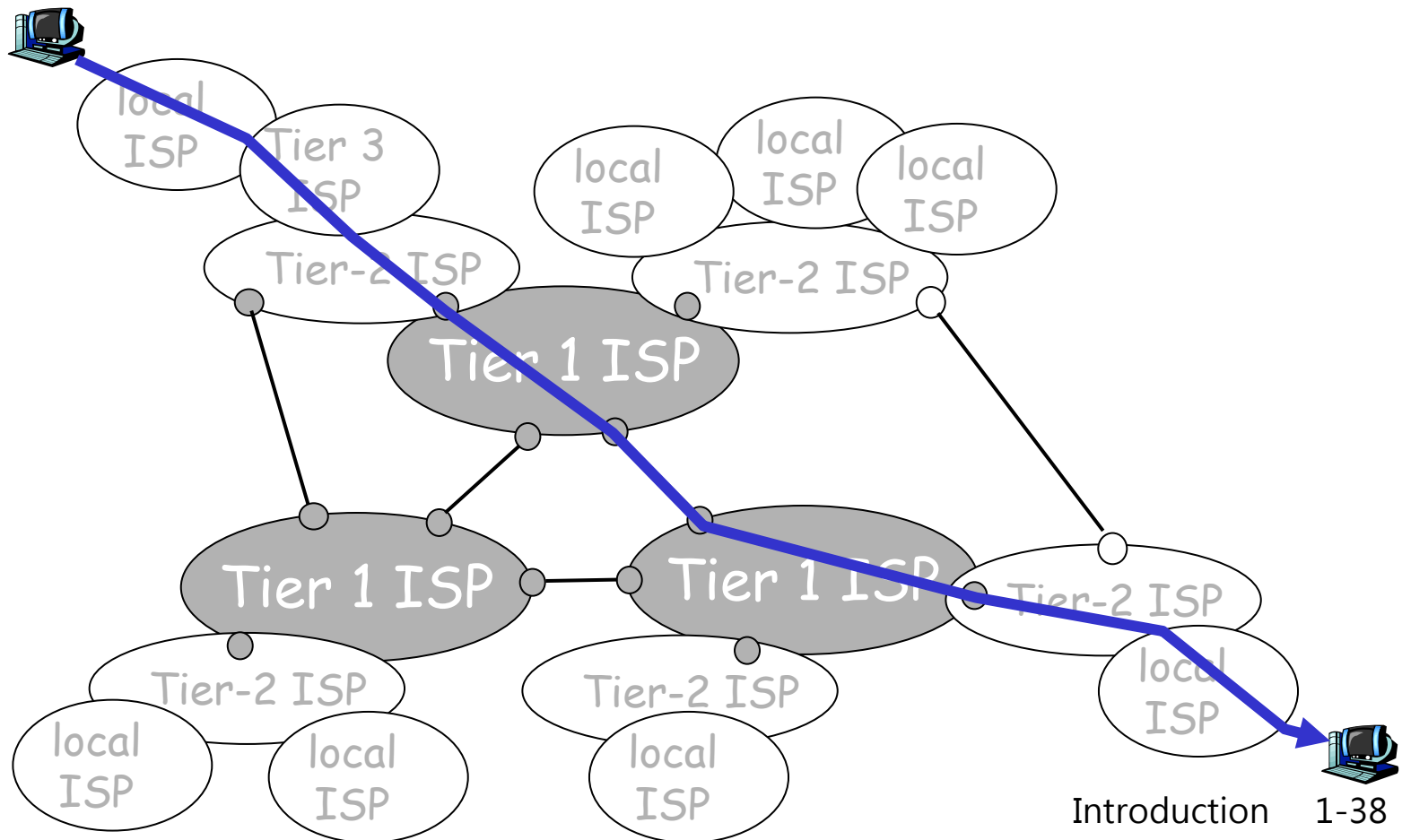
## ❑ "Tier-3" ISPs and local ISPs

- ❖ last hop ("access") network (closest to end systems)



# Internet structure: network of networks

- ❑ a packet passes through many networks!
- ❑ 透過多個網路傳輸封包



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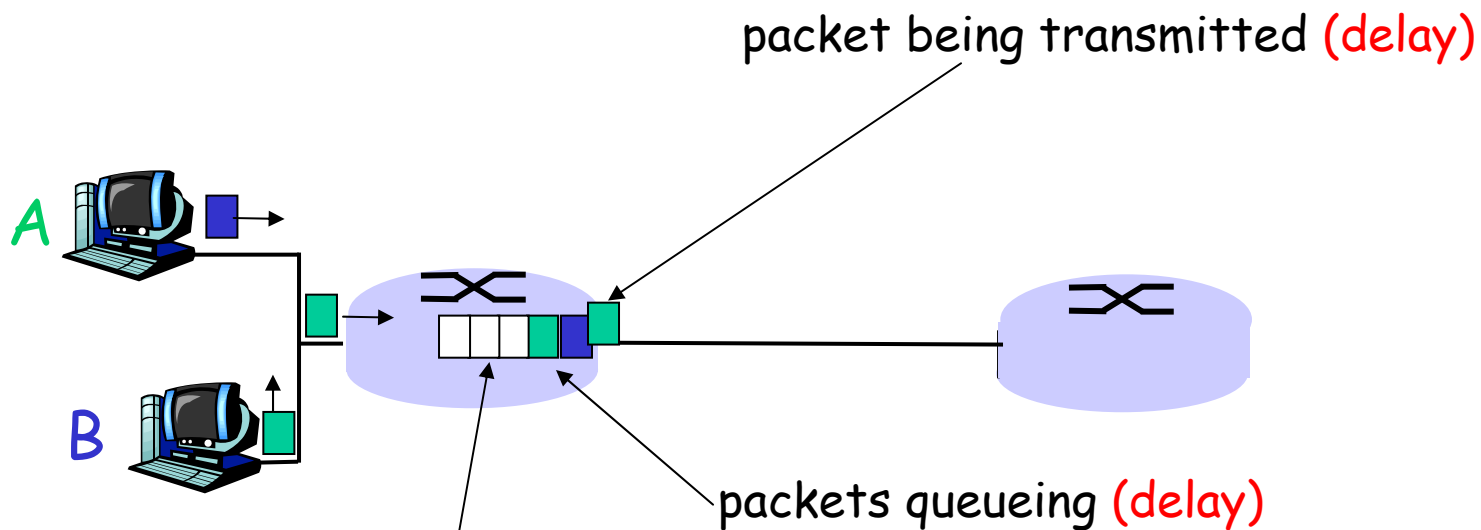
1.6 Networks under attack: security

1.7 History

# How do loss 遺失 and delay 延遲 occur?

packets *queue* in router buffers

- packet arrival rate to link exceeds output link capacity 到達速率大於處理速率
- packets queue, wait for turn 封包排隊



free (available) buffers: arriving packets  
dropped (**loss**) if no free buffers 緩衝區大小限制

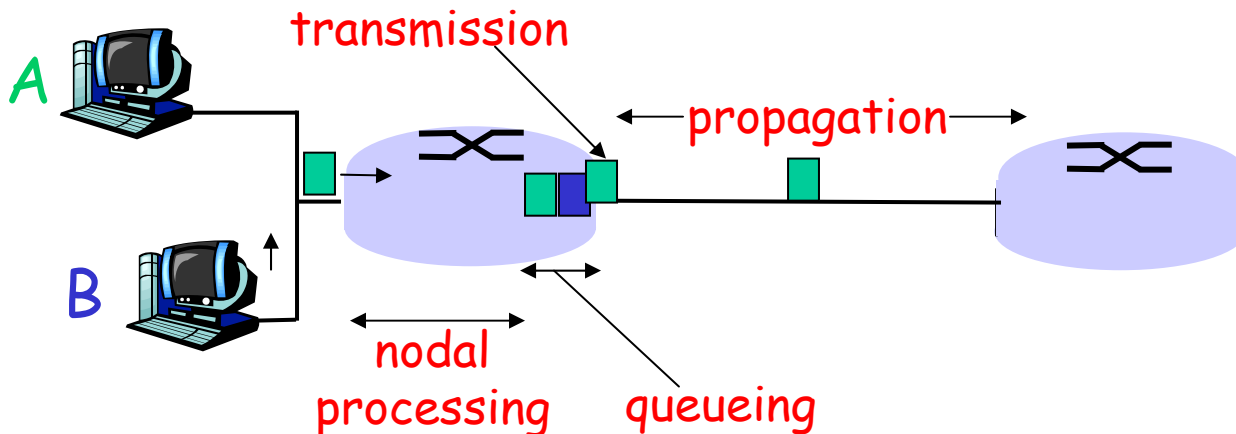


# Four sources of packet delay

## 封包延遲的四個來源

- 1. nodal processing:  
節點的訊號處理
  - ❖ check bit errors
  - ❖ determine output link

- 2. queueing  
等待被處理的時間
  - ❖ time waiting at output link for transmission
  - ❖ depends on congestion level of router



# Delay in packet-switched networks

## 3. Transmission delay: 傳輸延遲

(由機器送至傳輸媒介上)

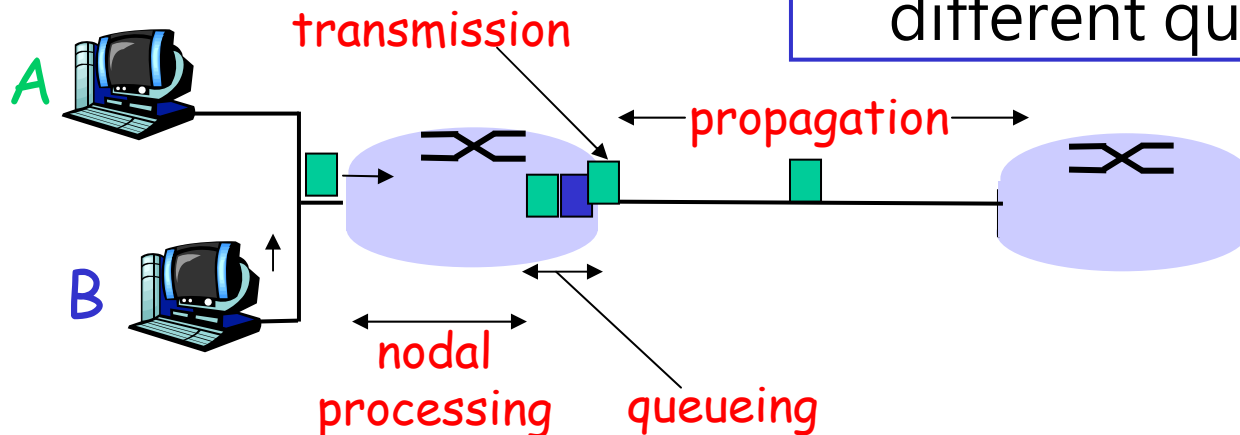
- $R$  = link bandwidth (bps)
- $L$  = packet length (bits)
- time to send bits into link =  $L/R$

## 4. Propagation delay: 傳播延遲

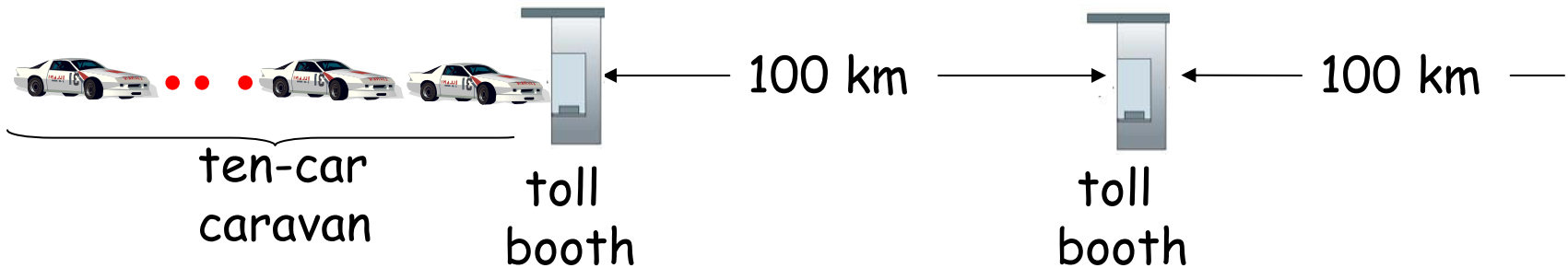
(在傳輸媒介上傳輸的時間)

- $d$  = length of physical link
- $s$  = propagation speed in medium ( $\sim 2 \times 10^8$  m/sec)  
光速或電子傳輸速度
- propagation delay =  $d/s$

**Note:**  $s$  and  $R$  are *very* different quantities!

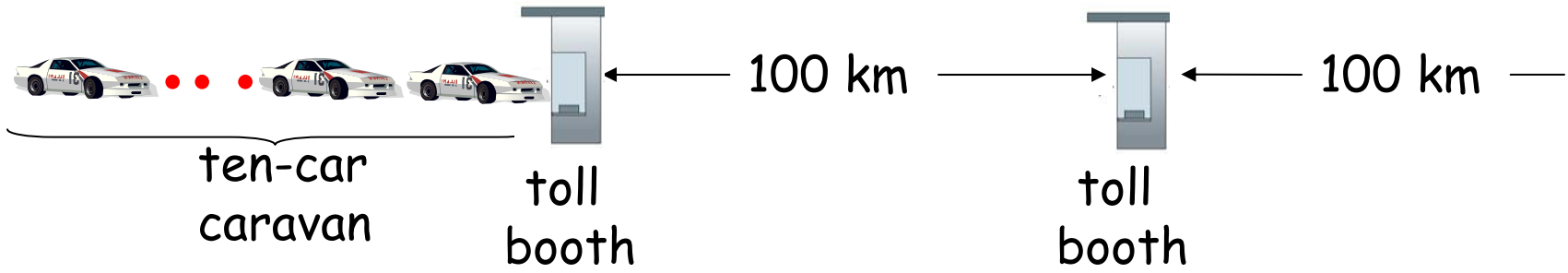


# Caravan analogy 以車隊來類比



- ❑ cars "propagate" at 100 km/hr 車速
- ❑ toll booth takes 12 sec to service car (transmission time) 每12秒處理一輛車
- ❑ car~bit; caravan ~ packet  
車=bit ; 車隊=封包
- ❑ Q: How long until caravan is lined up before 2nd toll booth?
- ❑ Time to "push" entire caravan through toll booth onto highway =  $12 \times 10 = 120$  sec
- ❑ Time for last car to propagate from 1st to 2nd toll booth:  
 $100\text{km} / (100\text{km/hr}) = 1$  hr
- ❑ A: 62 minutes

# Caravan analogy (more)



- ❑ Cars now “propagate” at 1000 km/hr
- ❑ Toll booth now takes 1 min to service a car
- ❑ **Q: Will cars arrive to 2nd booth before all cars serviced at 1st booth?**

- ❑ **Yes!** After 7 min, 1st car at 2nd booth and 3 cars still at 1st booth.
- ❑ 1st bit of packet can arrive at 2nd router before packet is fully transmitted at 1st router!
  - ❖ See Ethernet applet at AWL Web site

# Nodal delay

$$d_{\text{nodal}} = d_{\text{proc}} + d_{\text{queue}} + d_{\text{trans}} + d_{\text{prop}}$$

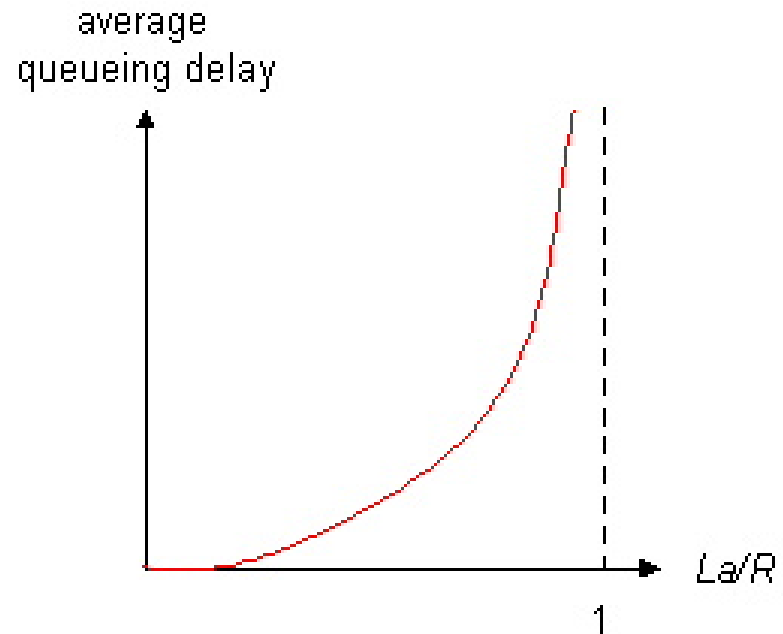
- $d_{\text{proc}}$  = processing delay
  - ❖ typically a few microseconds or less
- $d_{\text{queue}}$  = queuing delay 和流量有關
  - ❖ depends on congestion
- $d_{\text{trans}}$  = transmission delay 和封包大小、頻寬有關
  - ❖  $= L/R$ , significant for low-speed links
- $d_{\text{prop}}$  = propagation delay 和傳輸媒體、傳輸距離有關
  - ❖ a few microseconds to hundreds of msecs

# Queueing delay (revisited)

- $R$  = link bandwidth (bps)
- $L$  = packet length (bits)
- $a$  = average packet arrival rate

traffic intensity =  $\lambda a / R$

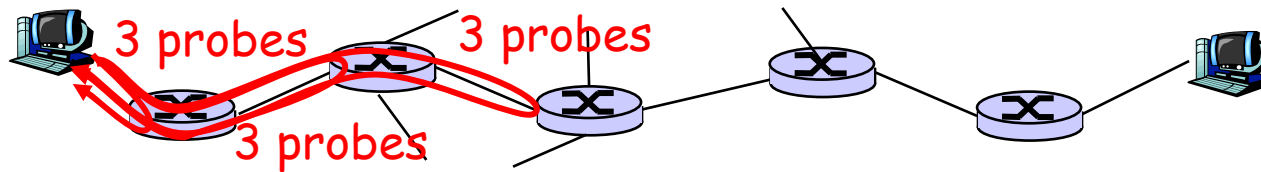
流量強度的定義



- $\lambda a / R \sim 0$ : average queueing delay small
- $\lambda a / R \rightarrow 1$ : delays become large
- $\lambda a / R > 1$ : more "work" arriving than can be serviced, average delay **infinite!**

# “Real” Internet delays and routes


- ❑ What do “real” Internet delay & loss look like?
- ❑ Traceroute program: provides delay measurement from source to router along end-end Internet path towards destination. For all  $i$ :
  - ❖ sends three packets that will reach router  $i$  on path towards destination
  - ❖ router  $i$  will return packets to sender
  - ❖ sender times interval between transmission and reply.



# “Real” Internet delays and routes

**traceroute:** gaia.cs.umass.edu to www.eurecom.fr

Three delay measurements from  
gaia.cs.umass.edu to cs-gw.cs.umass.edu



1 cs-gw (128.119.240.254) 1 ms 1 ms 2 ms  
2 border1-rt-fa5-1-0.gw.umass.edu (128.119.3.145) 1 ms 1 ms 2 ms  
3 cht-vbns.gw.umass.edu (128.119.3.130) 6 ms 5 ms 5 ms  
4 jn1-at1-0-0-19.wor.vbns.net (204.147.132.129) 16 ms 11 ms 13 ms  
5 jn1-so7-0-0-0.wae.vbns.net (204.147.136.136) 21 ms 18 ms 18 ms  
6 abilene-vbns.abilene.ucaid.edu (198.32.11.9) 22 ms 18 ms 22 ms  
7 nycm-wash.abilene.ucaid.edu (198.32.8.46) 22 ms 22 ms 22 ms  
8 62.40.103.253 (62.40.103.253) 104 ms 109 ms 106 ms  
9 de2-1.de1.de.geant.net (62.40.96.129) 109 ms 102 ms 104 ms  
10 de.fr1.fr.geant.net (62.40.96.50) 113 ms 121 ms 114 ms  
11 renater-gw.fr1.fr.geant.net (62.40.103.54) 112 ms 114 ms 112 ms  
12 nio-n2.cssi.renater.fr (193.51.206.13) 111 ms 114 ms 116 ms  
13 nice.cssi.renater.fr (195.220.98.102) 123 ms 125 ms 124 ms  
14 r3t2-nice.cssi.renater.fr (195.220.98.110) 126 ms 126 ms 124 ms  
15 eurecom-valbonne.r3t2.ft.net (193.48.50.54) 135 ms 128 ms 133 ms  
16 194.214.211.25 (194.214.211.25) 126 ms 128 ms 126 ms  
17 \* \* \*  
18 \* \* \*  
19 fantasia.eurecom.fr (193.55.113.142) 132 ms 128 ms 136 ms

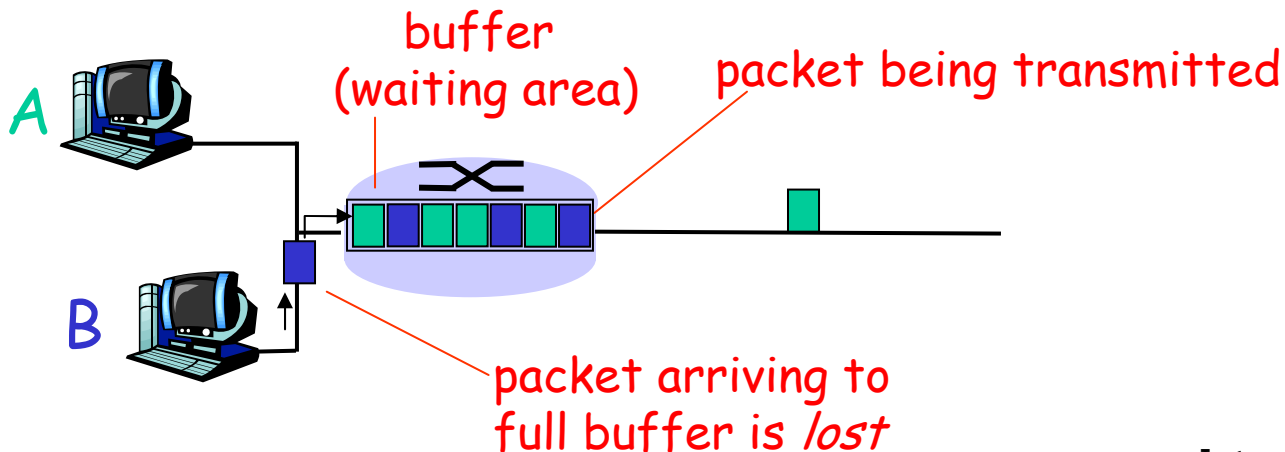
trans-oceanic link

\* means no response (probe lost, router not replying)



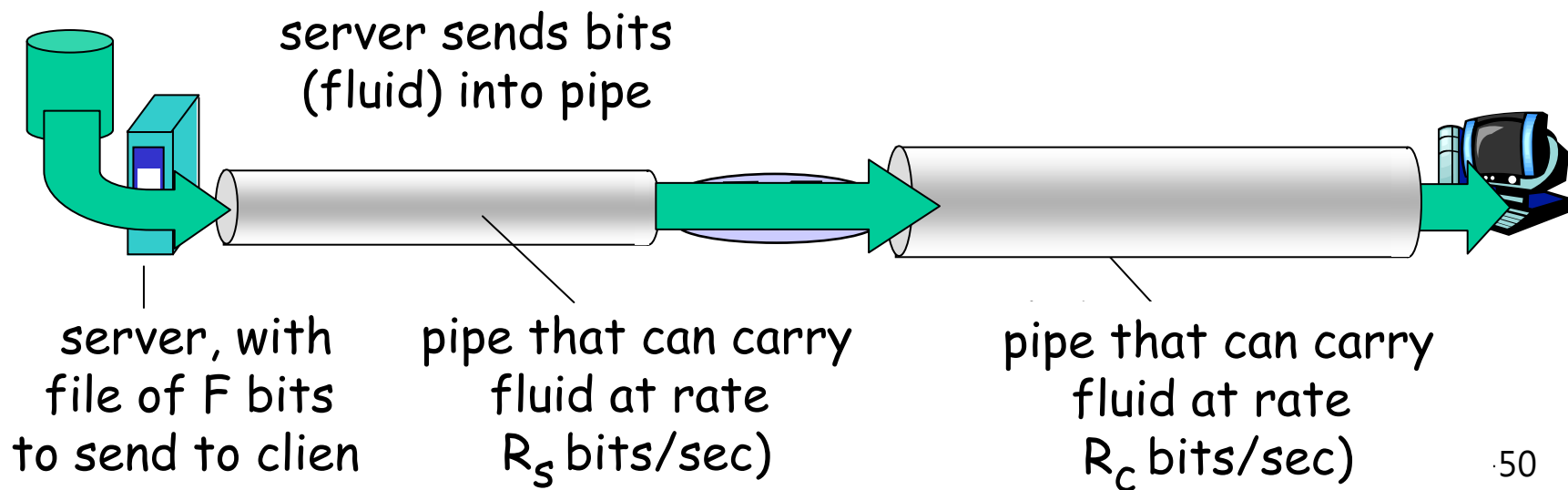
# Packet loss 封包遺失

- ❑ queue (aka buffer) preceding link in buffer has finite capacity 緩衝區容量有限制
- ❑ packet arriving to full queue dropped (aka lost) 緩衝區沒有足夠容量時則丟棄封包
- ❑ lost packet may be retransmitted (重傳) by previous node, by source end system, or not at all



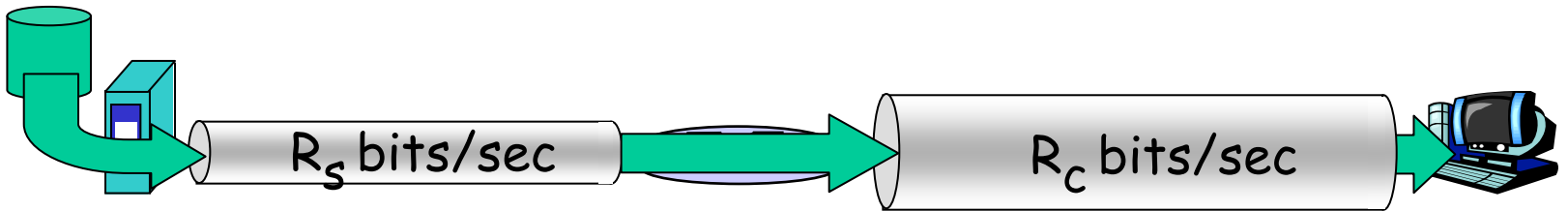
# Throughput 產出量

- *throughput*: rate (bits/time unit) at which bits transferred between sender/receiver  
每單位時間傳送的bit數量
  - ❖ *Instantaneous*: rate at given point in time 瞬間
  - ❖ *Average*: rate over long(er) period of time 平均

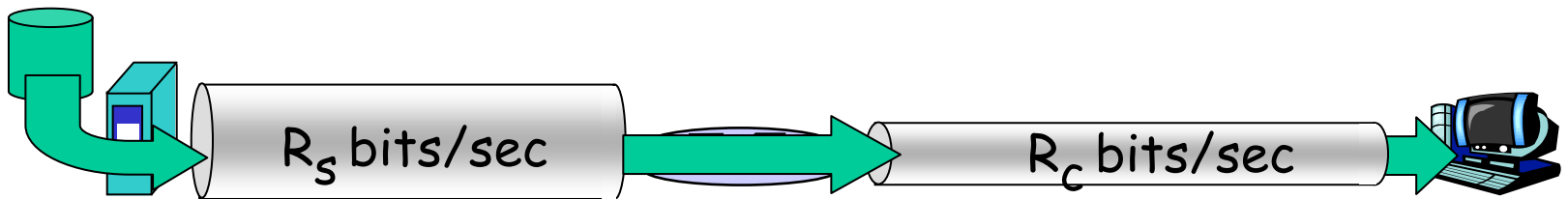


# Throughput (more) 瓶頸連結

- $R_s < R_c$  What is average end-end throughput?  $R_s$



- $R_s > R_c$  What is average end-end throughput?  $R_c$



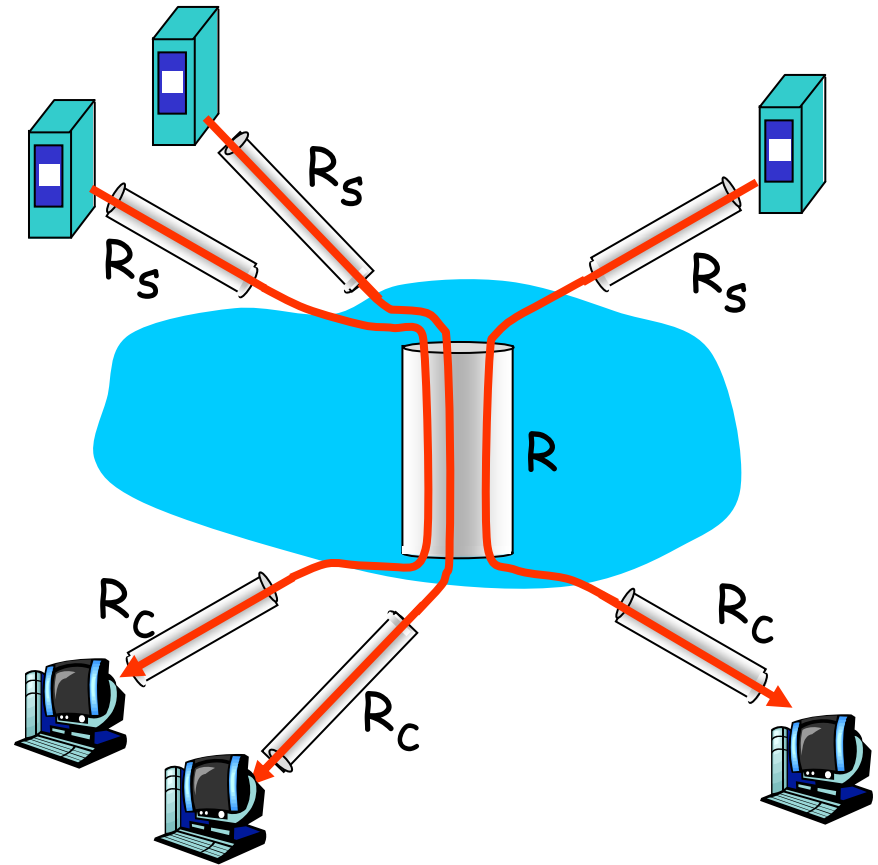
*bottleneck link*

link on end-end path that constrains end-end throughput =  $\min\{R_c, R_s\}$

# Throughput: Internet scenario

## 網際網路的實際狀況

- per-connection end-end throughput:  $\min(R_c, R_s, R/10)$   
 $R$  為 link 上的頻寬  
( 傳輸速率 )
- in practice:  $R_c$  or  $R_s$  is often bottleneck



10 connections (**fairly**) share backbone bottleneck link  $R$  bits/sec