# **Jaringan Komputer**

Pertemuan 2



Prodi Informatika

1

# **Outlines**

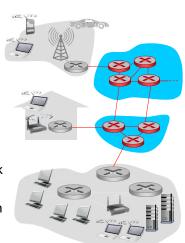
- Network Core (Packet Switching)
- Struktur Internet
- Delay, Lost dan Througput di Jaringan
- Protocol (Layer)
- Application Layer



Prodi Informatika

# The network core

- mesh network dari router yang saling terhubung
- packet-switching: host memecah pesan layer aplikasi ke dalam paket
  - Meneruskan paket dari satu router ke router berikutnya, melintasi Link di jalur dari sumber ke tujuan
  - setiap paket ditransmisikan dengan kapasitas link penuh

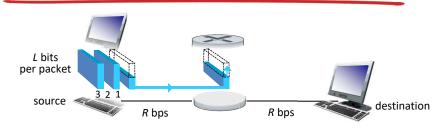






ರ

# Packet-switching: store-and-forward



- membutuhkan L / R detik untuk mengirimkan (mendorong keluar) paket L-bit ke dalam link di R bps
- store and forward: seluruh paket harus tiba di router sebelum dapat dikirim pada link berikutnya
- end-end delay = 2L/R (assuming zero propagation delay)

one-hop numerical example:

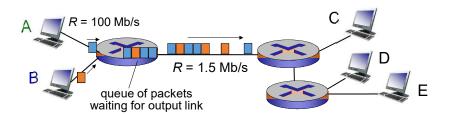
- L = 7.5 Mbits
- R = 1.5 Mbps
- one-hop transmission delay = 5 secmore on delay shortly ...

... ...

Introduction 1-4ka

Л

## Packet Switching: queueing delay, loss



#### queuing and loss:

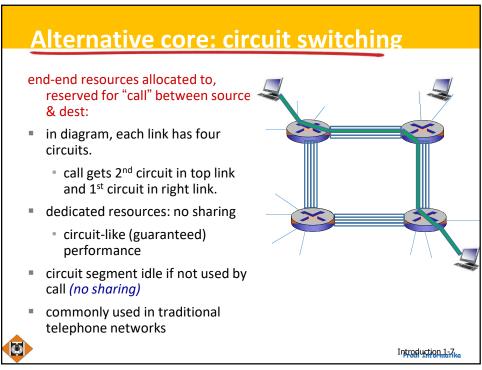
- jika tingkat kedatangan (dalam bit) untuk menghubungkan melebihi tingkat transmisi tautan untuk suatu periode waktu:
  - paket akan mengantri, menungg untuk dikirim pada link
  - · paket dapat didrop (hilang) jika memori (buffer) penuh



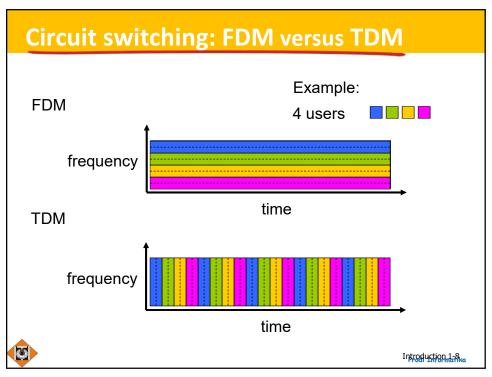
Introduction 1-5kg

5

## Two key network-core functions routing: determines sourcedestination route taken by forwarding: move packets packets from router's input to routing algorithms appropriate router output routing algorithm local forwarding table header value output link 0100 0101 3 2 2 1 0111 destination address in arriving packet's header Introduction 2-6kg



-

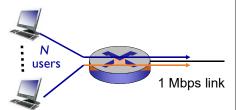


# Packet switching versus circuit switching

packet switching memungkinkan lebih banyak pengguna menggunakan jaringan!

#### example:

- 1 Mb/s link
- each user:
  - 100 kb/s when "active"
  - active 10% of time
- circuit-switching:
  - 10 users
- packet switching:
  - with 35 users, probability > 10 active at same time is less than .0004 \*



- Q: how did we get value 0.0004?
- Q: what happens if > 35 users?



Introduction 1-9kg

## Packet switching versus circuit switching

## is packet switching a "slam dunk winner?"

- great for bursty data
  - resource sharing
  - simpler, no call setup
- excessive congestion possible: 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
  - Q: human analogies of reserved resources (circuit switching) versus on-demand allocation (packet-switching)?



Introduction 1-10

# Internet structure: network of networks

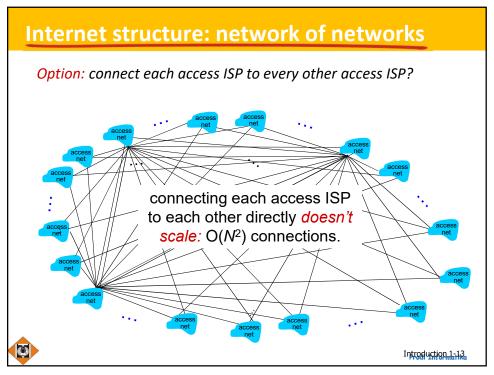
- End systems connect to Internet via access ISPs (Internet Service Providers)
  - residential, company and university ISPs
- Access ISPs in turn must be interconnected.
  - so that any two hosts can send packets to each other
- Resulting network of networks is very complex
  - evolution was driven by economics and national policies
- Let's take a stepwise approach to describe current Internet structure

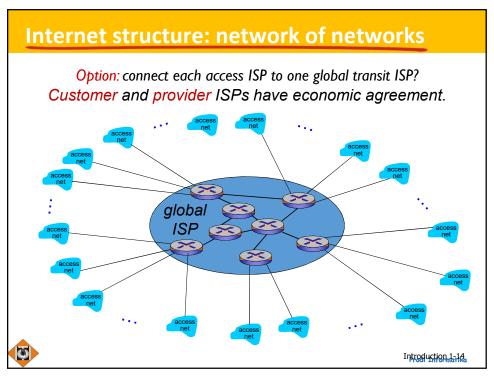


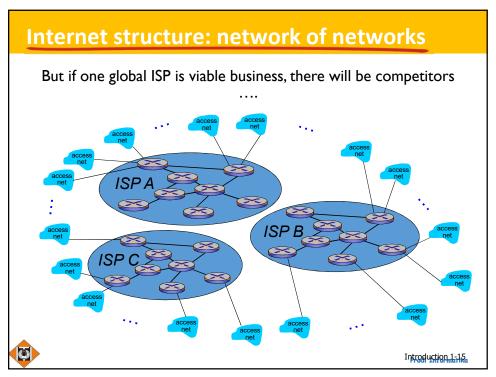
Introduction 1-11

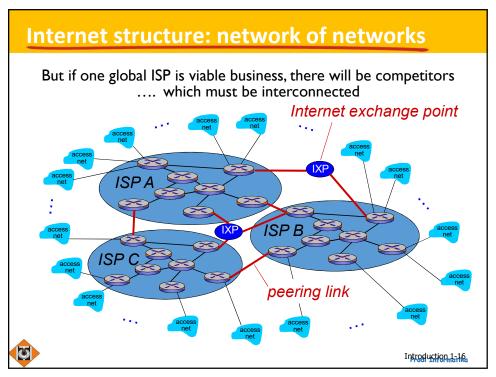
11

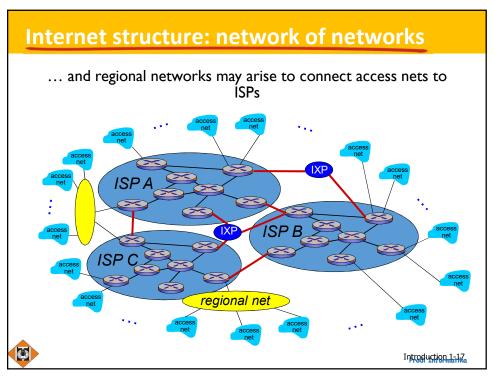
# Internet structure: network of networks Question: given millions of access ISPs, how to connect them together? | Cocess | Coces

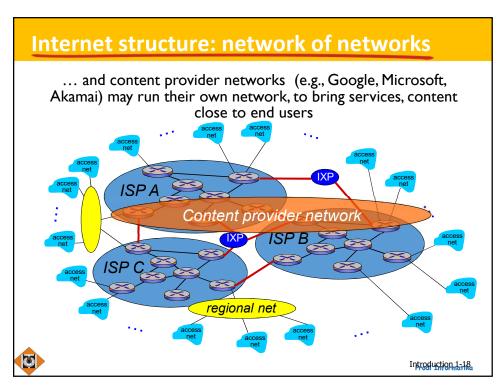


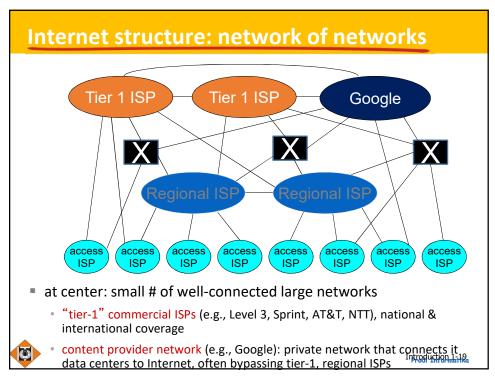


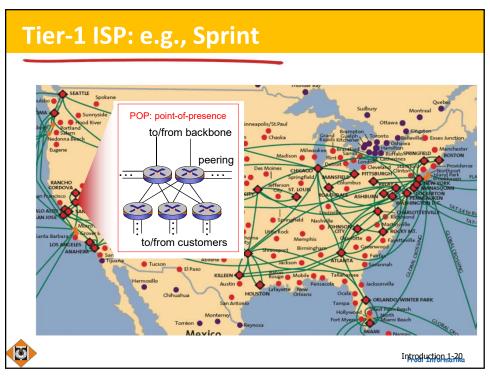












# Delay, loss, throughput in networks



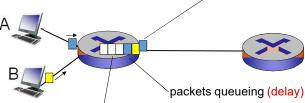
Introduction 1-21

21

# How do loss and delay occur?

## paket mengantri di buffer router

- tingkat kedatangan paket untuk link (sementara) melebihi kapasitas link keluaran
- paket antrian, tunggu giliran packet being transmitted (delay)

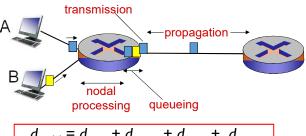


free (available) buffers: arriving packets dropped (loss) if no free buffers

O

Introduction 1-22





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

## $d_{proc}$ : nodal processing

- check bit errors
- determine output link
- typically < msec</p>

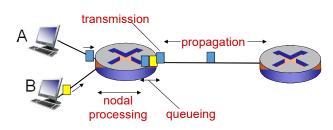
# d<sub>queue</sub>: queueing delay

- time waiting at output link for transmission
- depends on congestion level of router

Introduction 1-23

23

# Four sources of packet delay



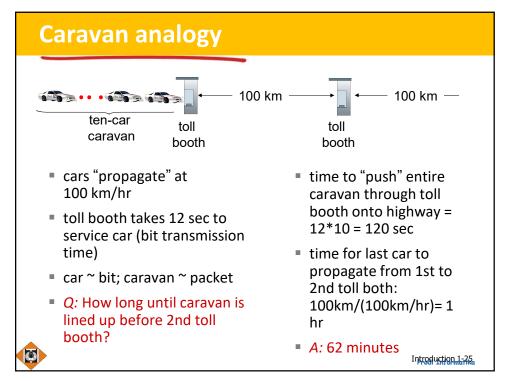
 $d_{\text{nodal}} = d_{\text{proc}} + d_{\text{queue}} + d_{\text{trans}} + d_{\text{prop}}$ 

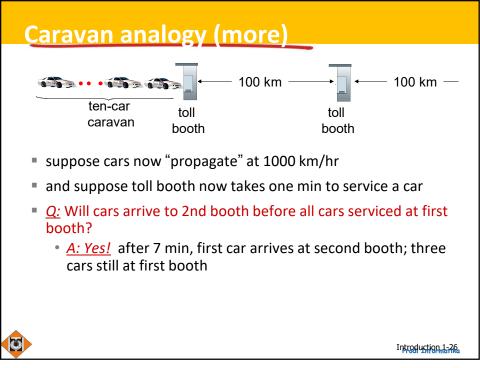
 $d_{trans}$ : transmission delay:

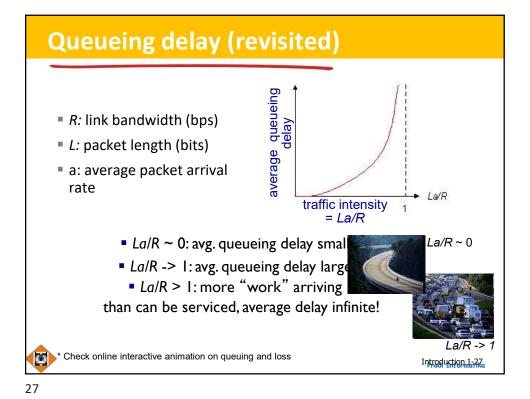
- $d_{prop}$ : propagation delay:
- L: packet length (bits)
- d: length of physical link
- R: link bandwidth (bps)
- s: propagation speed (~2x10<sup>8</sup> m/sec)
- $d_{trans} = L/R \leftarrow d_{trans}$  and  $d_{prop}$ 
  very different
- $\rightarrow$   $d_{prop} = d/s$



Introduction 1-24

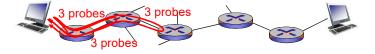






# "Real" Internet delays and routes

- 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.





Introduction 1-28

# "Real" Internet delays, routes

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

```
3 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 in1-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 (198.32.11.9) 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 (62.40.96.50) 113 ms 121 ms 114 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

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 ***

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

19 fantasia.eurecom.fr (193.55.113.142) 132 ms 128 ms 136 ms
```

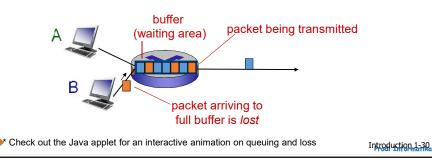
\* Do some traceroutes from exotic countries at www.traceroute.org

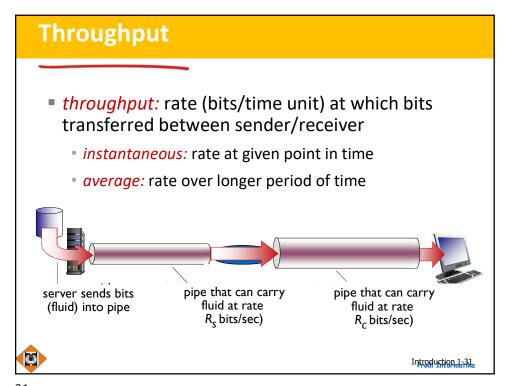
Introduction 1-29

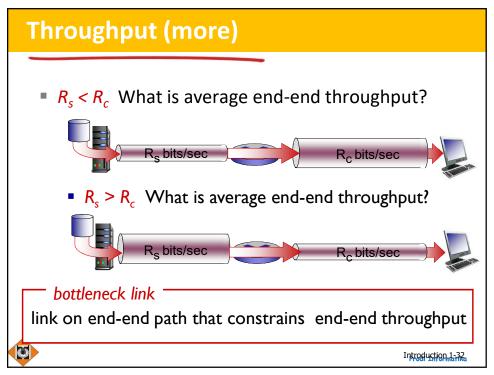
29

# **Packet loss**

- queue (aka buffer) preceding link in buffer has finite capacity
- packet arriving to full queue dropped (aka lost)
- lost packet may be retransmitted by previous node, by source end system, or not at all

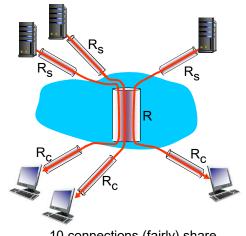




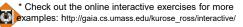


# **Throughput: Internet scenario**

- per-connection end-end throughput: min(R<sub>c</sub>, R<sub>s</sub>, R/10)
- in practice: R<sub>c</sub> or R<sub>s</sub> is often bottleneck



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



Introduction 1-33

33

# Protocol "layers"

# Networks are complex, with many "pieces":

- hosts
- routers
- links of various media
- applications
- protocols
- hardware, software

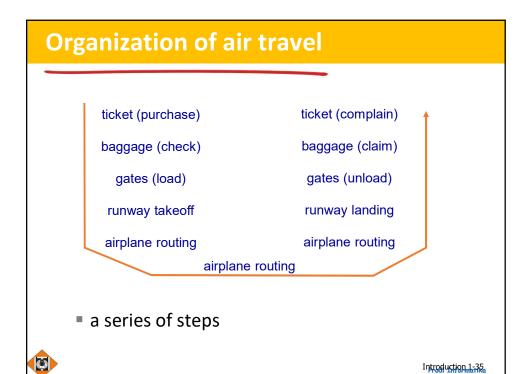
#### Question:

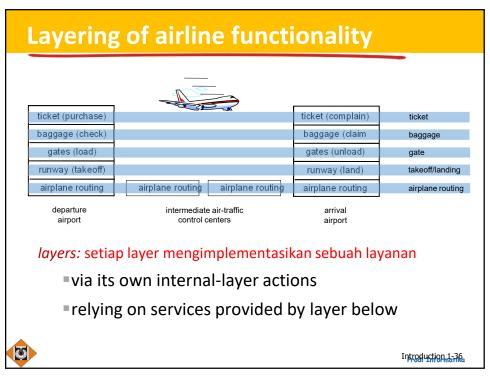
is there any hope of organizing structure of network?

.... or at least our discussion of networks?



Introduction 1-34





# Why layering?

dealing with complex systems:

- explicit structure allows identification, relationship of complex system's pieces
  - layered reference model for discussion
- modularization eases maintenance, updating of system
  - change of implementation of layer's service transparent to rest of system
  - e.g., change in gate procedure doesn't affect rest of system
- layering considered harmful?



Introduction 1-37

37

# Internet protocol stack

- application: supporting network applications
  - FTP, SMTP, HTTP
- transport: process-process data transfer
  - TCP, UDP
- network: routing of datagrams from source to destination
  - IP, routing protocols
- link: data transfer between neighboring network elements
  - Ethernet, 802.111 (WiFi), PPP

physical: bits "on the wire"

application

transport

network

link

physical

Introduction, 1:38.

# ISO/OSI reference model

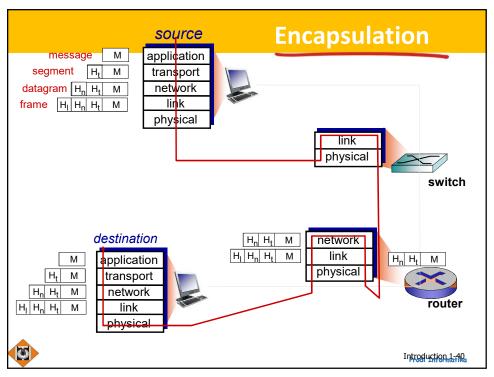
- presentation: allow applications to interpret meaning of data, e.g., encryption, compression, machine-specific conventions
- session: synchronization, checkpointing, recovery of data exchange
- Internet stack "missing" these layers!
  - these services, if needed, must be implemented in application
  - needed?

application
presentation
session
transport
network
link
physical

Introduction 1-39



39



# **Application layer**

## our goals:

- conceptual, implementation aspects of network application protocols
  - transport-layer service models
  - client-server paradigm
  - peer-to-peer paradigm
  - content distribution networks

- learn about protocols by examining popular application-level protocols
  - HTTP
  - FTP
  - SMTP / POP3 / IMAP
  - DNS
- creating network applications
  - socket API



Application Layer 2-41

# Some network apps

- e-mail
- web
- text messaging
- remote login
- P2P file sharing
- multi-user network games
- streaming stored video (YouTube, Hulu, Netflix)

- voice over IP (e.g., Skype)
- real-time video conferencing
- social networking
- search
- ...
- ..



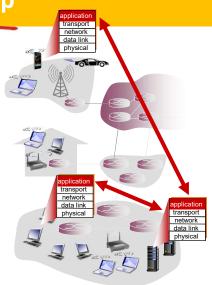
# **Creating a network app**

## write programs that:

- run on (different) end systems
- communicate over network
- e.g., web server software communicates with browser software

# no need to write software for network-core devices

- network-core devices do not run user applications
- applications on end systems allows for rapid app development, propagation







43

# **Application architectures**

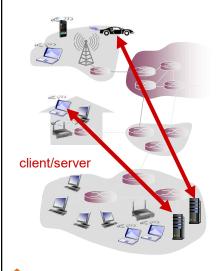
# possible structure of applications:

- client-server
- peer-to-peer (P2P)



Prodi Informatika

# **Client-server architecture**



#### server:

- always-on host
- permanent IP address
- data centers for scaling

#### clients:

- communicate with server
- may be intermittently connected
- may have dynamic IP addresses
- do not communicate directly with each other

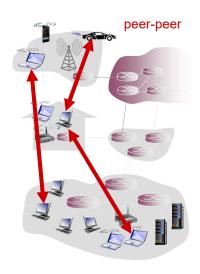
Prodi Informatika

# **P2P** architecture

- no always-on server
- arbitrary end systems directly communicate
- peers request service from other peers, provide service in return to other peers
  - self scalability new peers bring new service capacity, as well as new service demands
- peers are intermittently connected and change IP addresses



complex management



Prodi Informatika

# **Processes communicating**

# process: program running within a host

- within same host, two processes communicate using inter-process communication (defined by OS)
- processes in different hosts communicate by exchanging messages

## clients, servers

client process: process that initiates communication

server process: process that waits to be contacted

 aside: applications with P2P architectures have client processes & server processes

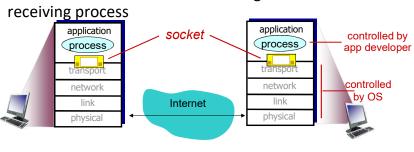


Application Layer 2-47

47

# **Sockets**

- process sends/receives messages to/from its socket
- socket analogous to door
  - sending process shoves message out door
  - sending process relies on transport infrastructure on other side of door to deliver message to socket at





Prodi Informatika

## **Addressing processes**

- to receive messages, process must have identifier
- host device has unique 32bit IP address
- Q: does IP address of host on which process runs suffice for identifying the process?
  - A: no, many processes can be running on same host

- identifier includes both IP address and port numbers associated with process on host.
- example port numbers:

HTTP server: 80mail server: 25

to send HTTP message to gaia.cs.umass.edu web server:

• IP address: 128.119.245.12

• port number: 80

more shortly...

Application Layer 2-49



**1**α

# **App-layer protocol defines**

- types of messages exchanged,
  - e.g., request, response
- message syntax:
  - what fields in messages & how fields are delineated
- message semantics
  - meaning of information in fields
- rules for when and how processes send & respond to messages

### open protocols:

- defined in RFCs
- allows for interoperability
- e.g., HTTP, SMTP

#### proprietary protocols:

e.g., Skype



## What transport service does an app need?

## data integrity

- some apps (e.g., file transfer, web transactions) require
   100% reliable data transfer
- other apps (e.g., audio) can tolerate some loss timing
- some apps (e.g., Internet telephony, interactive games) require low delay to be "effective"

## throughput

- some apps (e.g., multimedia) require minimum amount of throughput to be "effective"
- other apps ("elastic apps")
   make use of whatever
   throughput they get

## security

encryption, data integrity,...



Application Layer 2-51

51

## Transport service requirements: common apps

| appl           | ication  | data loss     | throughput         | time sensitive     |
|----------------|----------|---------------|--------------------|--------------------|
|                |          |               |                    |                    |
| file           | transfer | no loss       | elastic            | no                 |
|                | e-mail   | no loss       | elastic            | no                 |
| Web doc        | uments   | no loss       | elastic            | no                 |
| real-time audi | io/video | loss-tolerant | audio: 5kbps-1Mbps | yes, 100's         |
|                |          |               | video:10kbps-5Mbps | msec               |
| stored audi    | io/video | loss-tolerant | same as above      |                    |
| interactive    | games    | loss-tolerant | few kbps up        | yes, few secs      |
| text mes       | ssaging  | no loss       | elastic            | yes, 100's         |
|                |          |               |                    | msec<br>ves and no |



## **Internet transport protocols services**

#### TCP service:

- reliable transport between sending and receiving process
- flow control: sender won't overwhelm receiver
- congestion control: throttle sender when network overloaded
- does not provide: timing, minimum throughput guarantee, security
- connection-oriented: setup required between client and server processes

#### **UDP** service:

- unreliable data transfer between sending and receiving process
- does not provide: reliability, flow control, congestion control, timing, throughput guarantee, security, or connection setup,
- Q: why bother? Why is there a UDP?



Application Layer 2-53

53

## Internet apps: application, transport protocols

| _      | application      | application layer protocol              | underlying<br>transport protocol |
|--------|------------------|---|----------------------------------|
|        | e-mail           | SMTP [RFC 2821]                         | TCP                              |
| remote | terminal access  | Telnet [RFC 854]                        | TCP                              |
|        | Web              | HTTP [RFC 2616]                         | TCP                              |
|        | file transfer    | FTP [RFC 959]                           | TCP                              |
| strear | ning multimedia  | HTTP (e.g., YouTube),<br>RTP [RFC 1889] | TCP or UDP                       |
| In     | ternet telephony | SIP, RTP, proprietary<br>(e.g., Skype)  | TCP or UDP                       |



# **Securing TCP**

## TCP & UDP

- no encryption
- cleartext passwds sent into socket traverse Internet in cleartext

#### SSL

- provides encrypted TCP connection
- data integrity
- end-point authentication

## SSL is at app layer

apps use SSL libraries, that "talk" to TCP

## SSL socket API

 cleartext passwords sent into socket traverse Internet encrypted

р



Application Layer 2-55