

Computer Networks Homework #7

December 13th, 2012

Transport Layer

Q1:

- What entities does the transport layer connect in contrast to the network layer?

Transport vs. network layer

A1:

- *Network layer:*
logical communication
between hosts.
- *Transport layer:*
logical communication
between processes

Multiplexing

Q2:

- Why is **multiplexing** and **demultiplexing** used for at the transport layer and what has the concept of **ports** to do with this?

Multiplexing

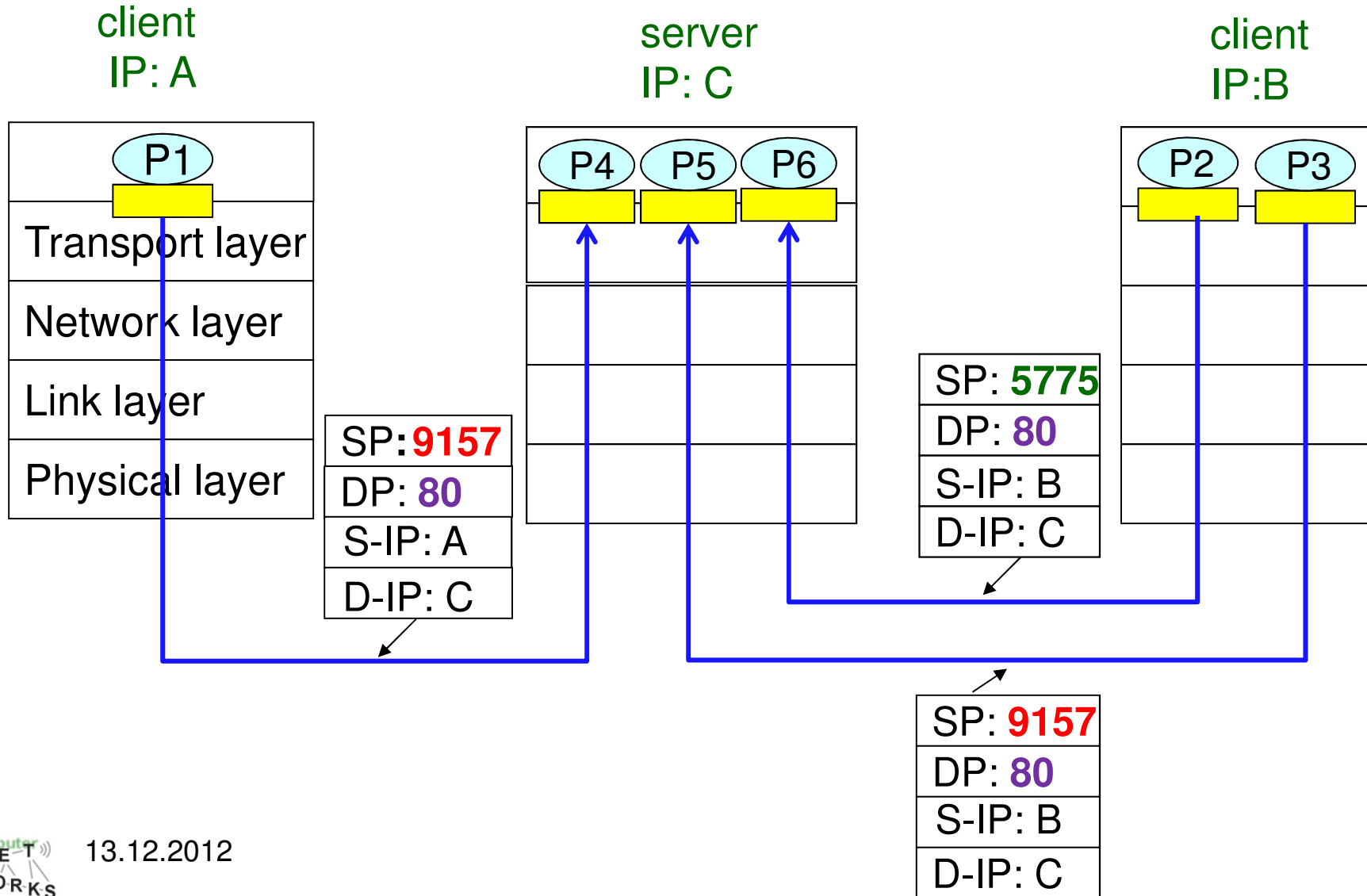
A2:

- **Demultiplexing:**
 - delivering the data in a transport-layer segment to the correct **socket**.
- **Multiplexing:**
 1. gathering data chunks at the source host from different **sockets**;
 2. encapsulating each data chunk with header information to create segments;
 3. passing the segments to the network layer.
- **Sockets:**

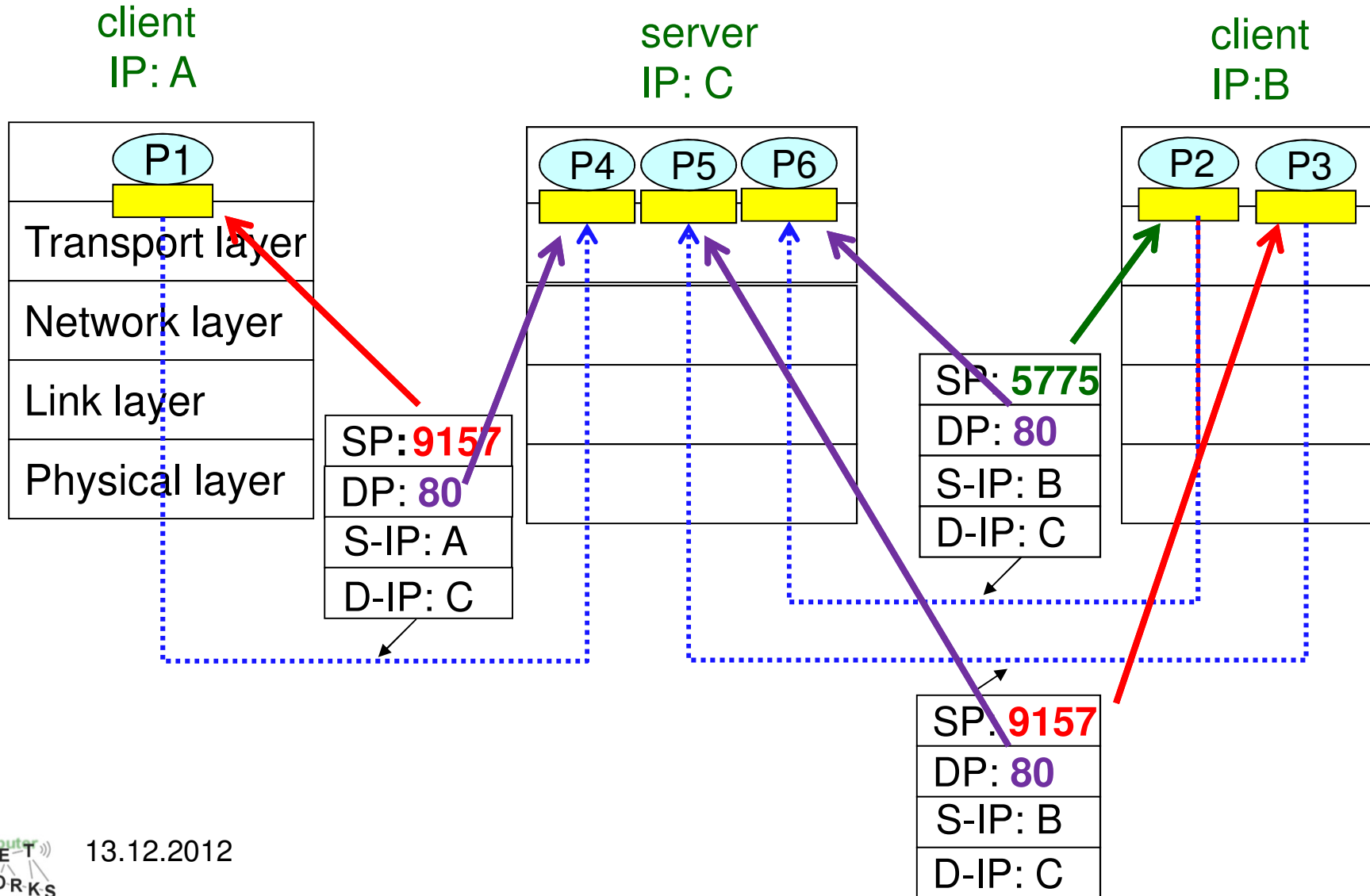
Doors

 - through which data passes from the network to the process;
 - through which data passes from the process to the network
- **Port Numbers:**
 - unique identifiers of sockets.

Multiplexing



Multiplexing



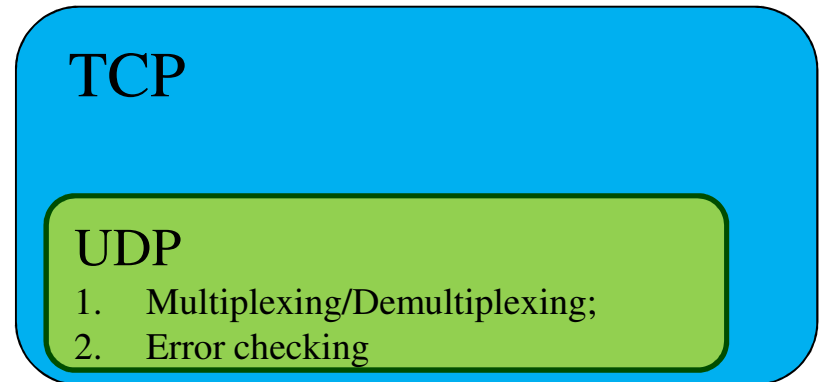
TCP vs. UDP

Q3:

- Please compare TCP and UDP in terms of the services they offer to the application layer.

TCP vs. UDP

A3:



- TCP

- Reliable data transfer
- Connection-oriented transport
- In-order delivery
- Congestion control

- UDP

- Best-effort data transfer
- Connectionless
- Possible out-of-order delivery
- No congestion control

TCP vs. UDP (II)

Q4:

- For which kinds of applications would you prefer UDP over TCP.

TCP vs. UDP (II)

A4:

- Transmission rate constraints enforced by TCP are not wanted/needed
- Reliability of TCP is not wanted/needed

Examples:

| Application | Application-Layer Protocol | Underlying Transport Protocol |
|------------------------|----------------------------|-------------------------------|
| Electronic mail | SMTP | TCP |
| Remote terminal access | Telnet | TCP |
| Web | HTTP | TCP |
| File transfer | FTP | TCP |
| Streaming multimedia | typically proprietary | UDP or TCP |
| Internet telephony | typically proprietary | UDP or TCP |
| Network management | SNMP | Typically UDP |
| Routing protocol | RIP | Typically UDP |
| Name translation | DNS | Typically UDP |

UDP checksums

Q5:

- Assume a UDP transport has received a datagram which consists of the following 16-bit words and the given checksum. Please verify the checksum.

- 1010 1010 1010 1010 (1st 16 bit word) **a1**
- 1011 1011 1011 1011 (2nd 16 bit word) **a2**
- 1100 1100 1100 1100 (3rd 16 bit word) **a3**
- 1110 1100 1100 1100 (recvd. checksum) **$2^{16} - (a1+a2+a3)$**

UDP checksums (cont'd)

A5:

- 1) one's complement sum of first two words:

$$\begin{array}{rcccccc} & 1010 & 1010 & 1010 & 1010 & \mathbf{a1} \\ + & 1011 & 1011 & 1011 & 1011 & \mathbf{a2} \\ \hline & 0110 & 0110 & 0110 & 0101 & \\ \hline & 0110 & 0110 & 0110 & 0110 & \mathbf{a12} \end{array}$$

Note: A red dashed box highlights the carry '1' from the first column, with a red arrow pointing to the right, indicating its addition to the last column's result.

- Why 0 at last digit? The carry is added in the last step!

UDP checksums (cont'd)

- 2) add third word:

$$\begin{array}{rcccc} & 0110 & 0110 & 0110 & 0110 & a1+a2 \\ + & 1100 & 1100 & 1100 & 1100 & a3 \\ \hline 1 & 0011 & 0011 & 0011 & 0011 & \\ \hline & 0110 & 0110 & 0110 & 0110 & a1+a2+a3 \end{array}$$

The diagram illustrates the addition of a third word to the running sum. The first two rows show the sum of the first two words (a1+a2) and the third word (a3). A carry of 1 is generated from the first column. This carry is added to the first column of the third row, resulting in a carry of 1 from the second column. The final result is the sum of all three words (a1+a2+a3).

UDP checksums (cont'd)

- 3) add to received checksum:

$$\begin{array}{r} 0011 \ 0011 \ 0011 \ 0011 \\ + 1110 \ 1100 \ 1100 \ 1100 \\ \hline 1 \ 0001 \ 1111 \ 1111 \ 1111 \end{array} \quad \begin{array}{l} \mathbf{a1+a2+a3} \\ \text{checksum: } 2^{16} - (a1+a2+a3) \\ = 2^{16} ? \end{array}$$

- 4) Result \neq 1111 1111 1111 1111 therefore verification failed

Reliable data transfer

Q6:

Assume you want to reliably transfer data over a channel with **bit errors** but **no loss**. An error detection mechanism is already implemented.

- Which **simple mechanism** can you use to recover from errors?
- What **flaw** does this simple mechanism have?

Reliable data transfer (cont'd)

A6:

rdt2.0: reliable data transfer 2.0

- Send **ACK** if packet was received without errors,
ACK: “OK!”
- Send **NAK** otherwise
NAK: “Please repeat that!”

- Flaw: If **ACK/NAK** gets corrupted, sender doesn't know if packet was received without errors.

Reliable data transfer (II)

Q7:

Assume you want reliably transfer data over a channel with **bit errors** and **loss**.

- What **additional mechanism** do you need to introduce?
- Give an **example** of how this mechanism can recover from the loss of a packet.

Reliable data transfer (II)

(cont'd)

A7:

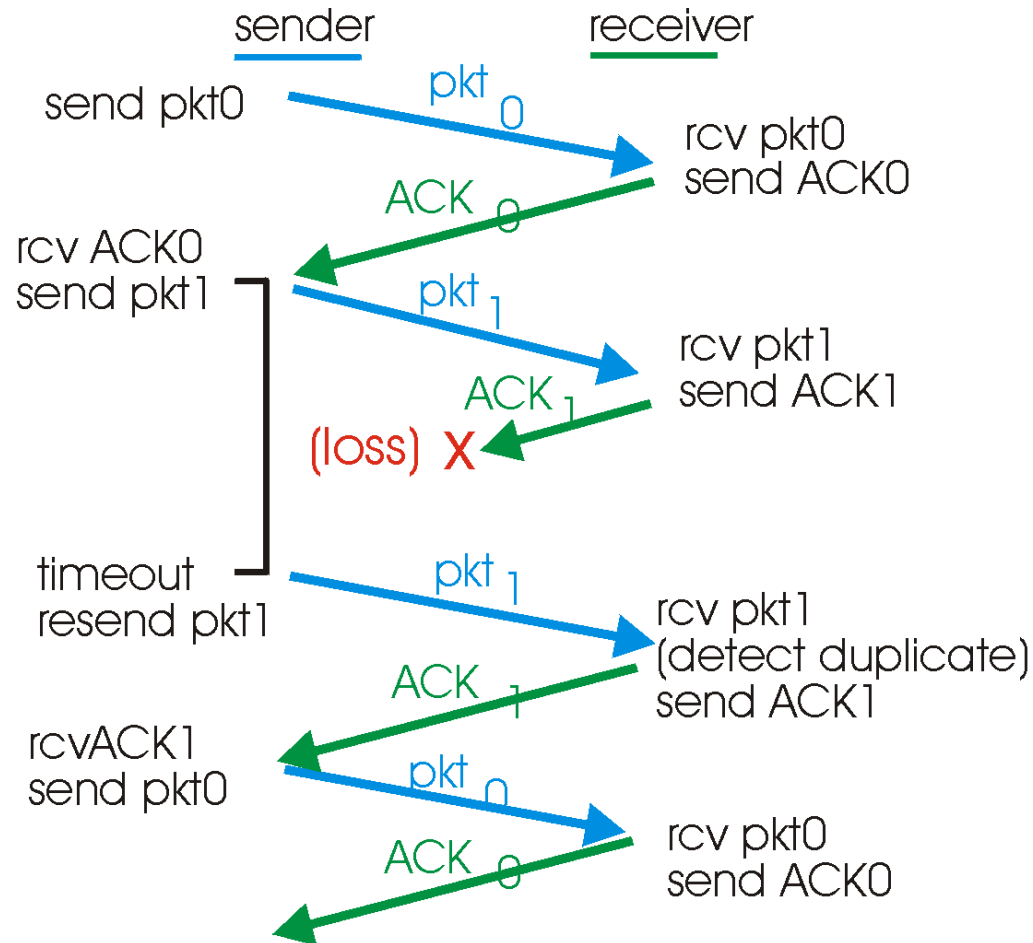
rdt3.0: reliable data transfer 3.0

- Informal:
 - Sender **wait** long enough **time** to *certain that a* packet has been lost, then **retransmit**.
- Formal:
 - Sender needs to maintain a timer for unacknowledged packets so it can re-transmit if an ACK is not received within a certain timeframe.

Reliable data transfer (II)

(cont'd)

Example:



- That's all and thanks for your attention!