



TECHNISCHE  
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**Technical University Darmstadt**

– Department of Computer Science –

## **Application-Layer Protocols on the Internet**

A Summary of the Course Contents

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## LIST OF ABBREVIATIONS

MAC	Medium Access Control
IP	Internet Protocol
HTTP	Hypertext Transfer Protocol
SSH	Secure Socket Shell
DNS	Domain Name System
DHCP	Dynamic Host Configuration Protocol
NTP	Network Time Protocol
NAT	Network Address Translation
RTT	Round Trip Time
POP <sub>3</sub>	Post Office Protocol version 3
IMAP	Internet Message Access Protocol
SMTP	Simple Mail Transfer Protocol
SSL	Secure Sockets Layer
TLS	Transport Layer Security
PPP	Point to Point Protocol
UDP	User Datagram Protocol
TCP	Transmission Control Protocol

## 1 FILESHARING, OVERLAYS AND ROBUSTNESS

## 2 BASIC TERMS AND DEFINITIONS

In this chapter, basic terms will be defined and explained.

### 1. **TCP/IP** Conceptual Layers:

- a) *Application*: Standardizes communication interfaces built on Transport Layer protocols for specific classes of applications.

**Protocols**: Hypertext Transfer Protocol (**HTTP**), Secure Socket Shell (**SSH**), Domain Name System (**DNS**), Dynamic Host Configuration Protocol (**DHCP**), Network Time Protocol (**NTP**), Post Office Protocol version 3 (**POP3**), Internet Message Access Protocol (**IMAP**), Simple Mail Transfer Protocol (**SMTP**), Secure Sockets Layer (**SSL**)/Transport Layer Security (**TLS**) (...)

- b) *Transport*: Provide process-to-process communication for application using ports. Features which might be given depending on the protocol are connection-oriented communication, reliability, error correction, flow-control and multiplexing (e.g. multicast / broadcast).

**Protocols**: Transmission Control Protocol (**TCP**), User Datagram Protocol (**UDP**) (...)

- c) *Internet*: Routing and transmission from source to destination.

**Protocols**: Internet Protocol (**IP**) (v4, v6) (...)

- d) *Network Interface*: Transmission of data between two computers in the same network via physical connection.

**Protocols**: Medium Access Control (**MAC**), Tunnels, Point to Point Protocol (**PPP**) (...)

- 2. **Datagram**: A datagram is a basic transfer unit in a packet-switched network. Datagrams consist of a header and payload. They contain address information and can be sent in connectionless communication.

- 3. **User Datagram Protocol (UDP)**: **UDP** provides a simple connectionless communication model. It provides checksums for verification of data integrity and port numbers for addressing specific services on the target machine. It does not have handshaking dialogues and requires no previous packets to be sent prior to communication. Messages are mapped directly to packets and are not split up into pieces. **UDP** does not provide a guarantee of delivery, order of packets, or duplicate protection. It therefore exposes the application layer to any unreliability of the network for the sake of less communication overhead. It is suitable for applications which do not require reliability but do require speed e.g. in video streaming or for applications which handle resubmission, ordering and duplicate checking themselves.

4. **Transmission Control Protocol (TCP):** TCP follows a connection-oriented communication model providing reliability, ordering and error checking and flow-control (not allowing one side to send too fast). Messages are handled as a stream of bytes which is split into packets of undefined size for transmission.

a) *Connection establishment and termination:* The connection is established using a three-way handshake consisting of the packets SYN  $\rightarrow$ , SYN-ACK  $\leftarrow$  and ACK  $\rightarrow$ . The connection termination is done with a four-way handshake consisting of the packets FIN  $\rightarrow$ , ACK  $\leftarrow$ , FIN  $\leftarrow$  and ACK  $\rightarrow$ . It is possible to shorten this by sending a FIN-ACK as a reply to the FIN

b) *Reliability:* TCP uses a sequence number to identify each byte of data, as shown in figure 1. The sequence number of the first byte is randomly chosen by the sender of the first packet in order to defend against TCP sequence prediction attacks. Each TCP-packet contains a sequence number and – if it is an ACK packet – an acknowledgement number. The sequence number of the packet is the sequence number of the first byte send in this message or the sequence number of the last byte send in previous messages in case the payload (data) of the packet has size zero (as for ACK packets). The acknowledgement number is the incremented sequence number of the last byte received. TCP uses cumulative ACKs, which means that an acknowledgement number of  $n$  acknowledges all bytes with sequence number  $< n$ .

c) *Resubmission:*

d)

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5. **Autonomous System:**

6. **Round Trip Time (RTT):** Is the transmission time for a packet from point A to point B plus the transmission time of the ACK-packet from point B back to point A. It can be determined using the ping command.

7. **Network Address Translation (NAT):**

8. **Subnetting:**

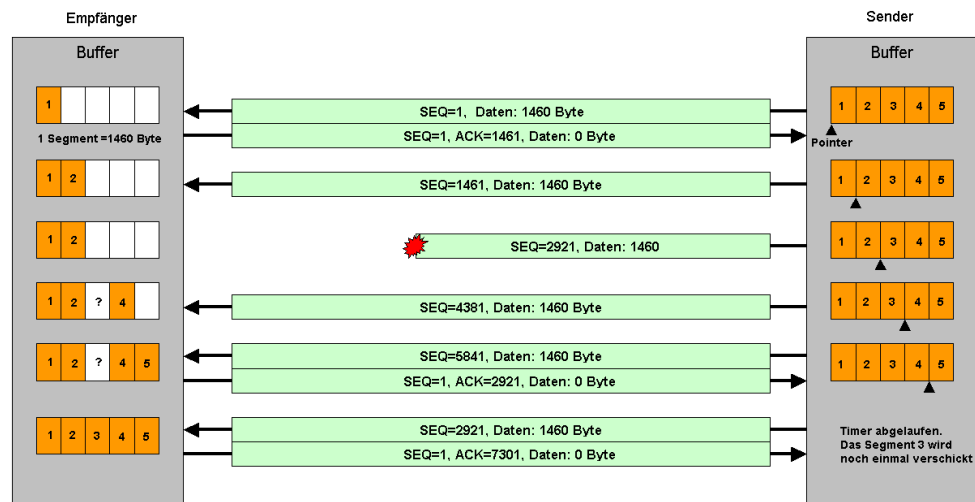


Figure 1: TCP data Transmission with sequence numbers