



## 2. Traffic

## Contents

- Traffic characterisation
- Telephone traffic modelling
- Data traffic modelling at packet level
- Data traffic modelling at flow level

## Offered vs. carried traffic

- **Offered** traffic
  - traffic as it is originally generated in the sources
- **Carried** traffic
  - traffic as it is carried by the network

## Characterisation of carried traffic

- Circuit-switched traffic
  - number of ongoing calls or active connections (erl)
  - may be converted into bit rate in digital systems
    - e.g. a telephone call reserves 64 kbps (=  $8000 \cdot 8$  bps) in a PCM system
- Packet-switched traffic
  - bit stream (bps, kbps, Mbps, Gbps, ...)
  - packet stream (pps)
  - number of active flows (erl)

## Traffic units

- Telephone traffic:
  - **erlangs** (erl)
  - one erlang corresponds to one ongoing call or one occupied channel
- Data traffic:
  - **bits per second** (bps)
  - packets per second (pps)
- Note:
  - 1 byte = 8 bits
  - 1 kbps = 1 kbit/s = 1,000 bits per second
  - 1 Mbps = 1 Mbit/s = 1,000,000 bits per second
  - 1 Gbps = 1 Gbit/s = 1,000,000,000 bits per second

## Traffic variations in different time scales (1)

- **Predictive** variations:
  - **Trend** (years)
    - traffic growth: due to
      - existing services (new users, new ways to use, new tariffs)
      - new services
  - Regular **year profile** (months)
  - Regular **week profile** (days)
  - Regular **day profile** (hours)
    - including “busy hour”
  - Variations caused by predictive (regular and irregular) **external events**
    - regular: e.g. Christmas day
    - irregular: e.g. televoting

## Traffic variations in different time scales (2)

- **Non-predictive variations:**
  - **Short term random variations** (seconds - minutes)
    - random call arrivals
    - random call holding times
  - **Long term random variations** (hours - ...)
    - random deviations around the profiles
    - each day, week, month, etc. is different
  - Variations caused by non-predictive **external events**
    - e.g. earthquakes and other natural disasters
- **Note:**
  - Ordinary traffic theoretic models focus on **short term random variations**

## Busy hour (1)

- For dimensioning,
  - an estimate of the traffic load is needed
- In telephone networks,
  - standard way is to use so called **busy hour** traffic for dimensioning

**Busy hour**  $\approx$  the continuous 1-hour period for which the traffic volume is greatest

- This is unambiguous only for a single day (let's call it **daily peak hour**)
- For dimensioning, however, we have to look at not only a single day but many more
- Different definitions for busy hour (covering several days) traffic have been proposed by ITU:
  - Average Daily Peak Hour (ADPH)
  - Time Consistent Busy Hour (TCBH)



## Busy hour (2)

- Let
  - $N$  = number of days during which measurements are done (e.g.  $N = 10$ )
  - $a_n(\Delta)$  = measured average traffic during 1-hour interval  $\Delta$  of day  $n$
  - $\max_{\Delta} a_n(\Delta)$  = daily peak hour traffic of day  $n$
- Busy hour traffic  $a$  with different methods:

$$a_{\text{ADPH}} = \frac{1}{N} \sum_{n=1}^N \max_{\Delta} a_n(\Delta)$$

$$a_{\text{TCBH}} = \max_{\Delta} \frac{1}{N} \sum_{n=1}^N a_n(\Delta)$$

- Note that

$$a_{\text{TCBH}} \leq a_{\text{ADPH}}$$

## Demo: Funet

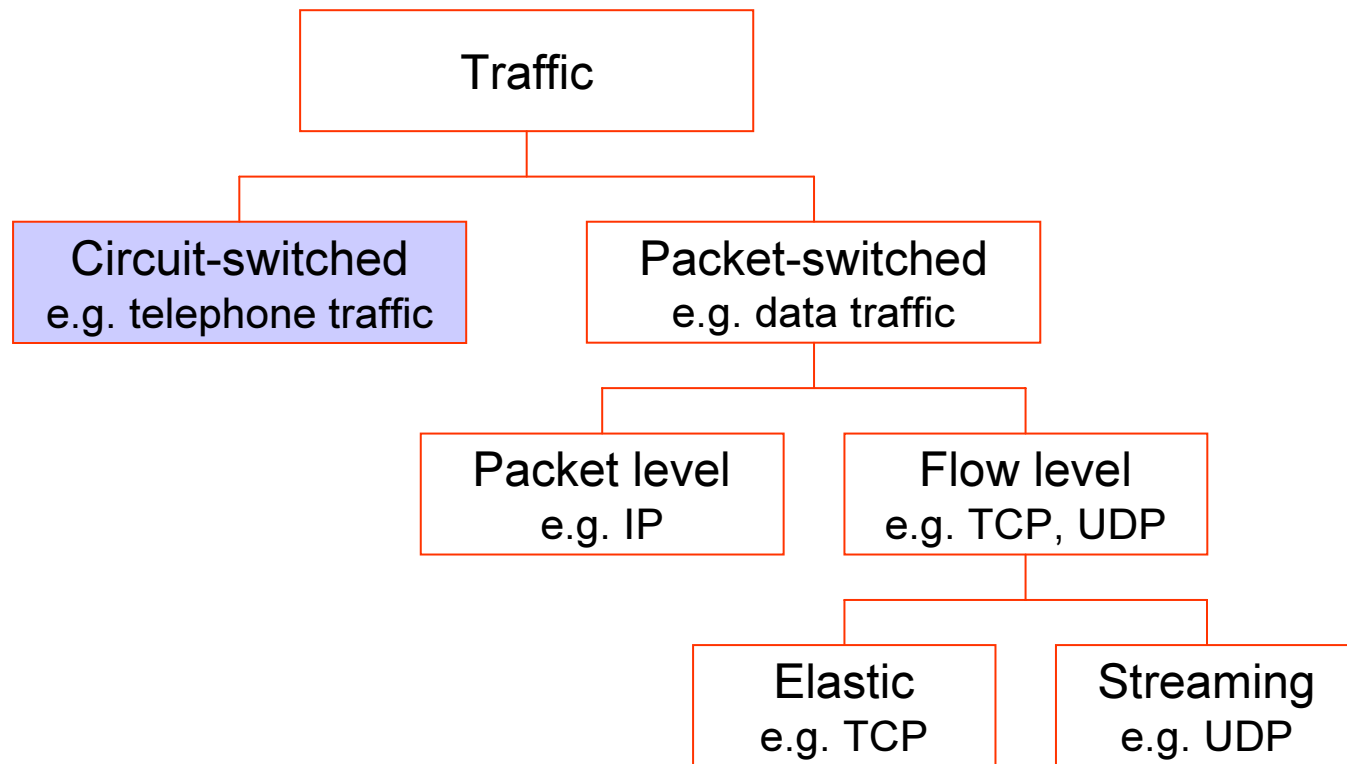
- Diurnal pattern, day profile
  - day vs. night
  - peak traffic, busy "hour"
  - changes in routing?
- Week profile
  - working days vs. weekend
- Month profile
  - special days: e.g. Christmas day
- Year profile
- Long-term trend?

```
http://www.csc.fi/suomi/funet/verkko.html.fi  
http://www.csc.fi/suomi/funet/noc/looking-glass/wm
```

## Contents

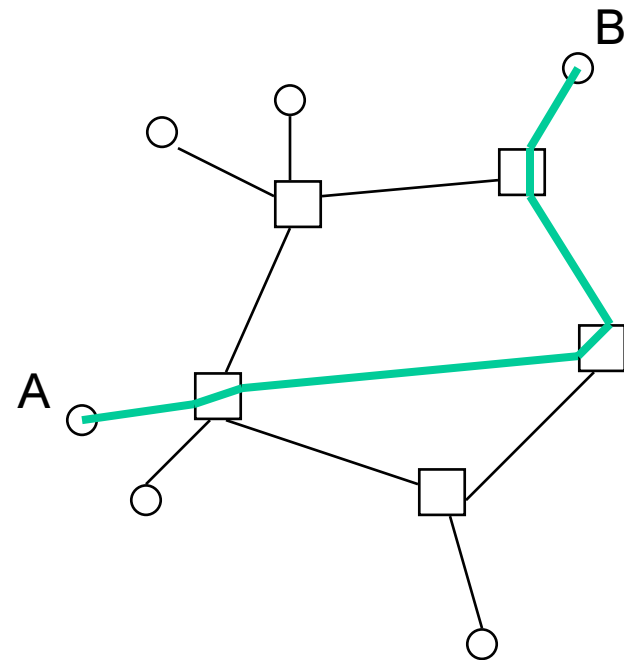
- Traffic characterisation
- Telephone traffic modelling
- Data traffic modelling at packet level
- Data traffic modelling at flow level

# Traffic classification



## Telephone network

- Connection oriented:
  - connections set up end-to-end before information transfer
  - resources reserved for the whole duration of connection
  - if resources are not available, the call is blocked and lost
- Information transfer as continuous stream

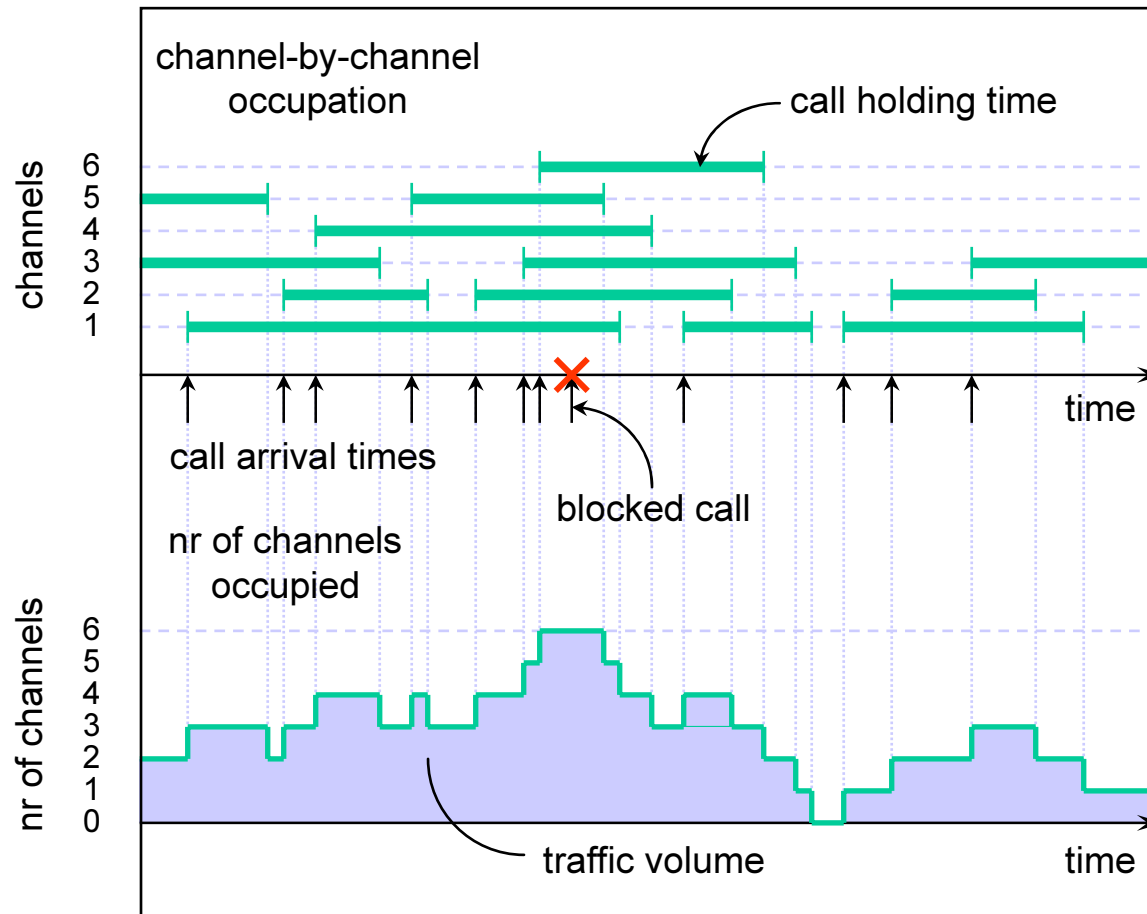


## Telephone traffic model

- Telephone traffic consists of **calls**
  - a call occupies one channel from each of the links along its route
  - call characterisation: **holding time** (in time units)
- Modelling of offered traffic:
  - **call arrival process** (at which moments new calls arrive)
  - **holding time distribution** (how long they take)
- Link model: a pure **loss system**
  - a server corresponds to a channel
  - the service rate  $\mu$  depends on the **average holding time**
  - the number of servers,  $n$ , depends on the **link capacity**
  - when all channels are occupied, **call admission control** rejects new calls so that they will be blocked and lost
- Modelling of carried traffic:
  - **traffic process** tells the number of ongoing calls = the number of occupied channels

## 2. Traffic

# Traffic process

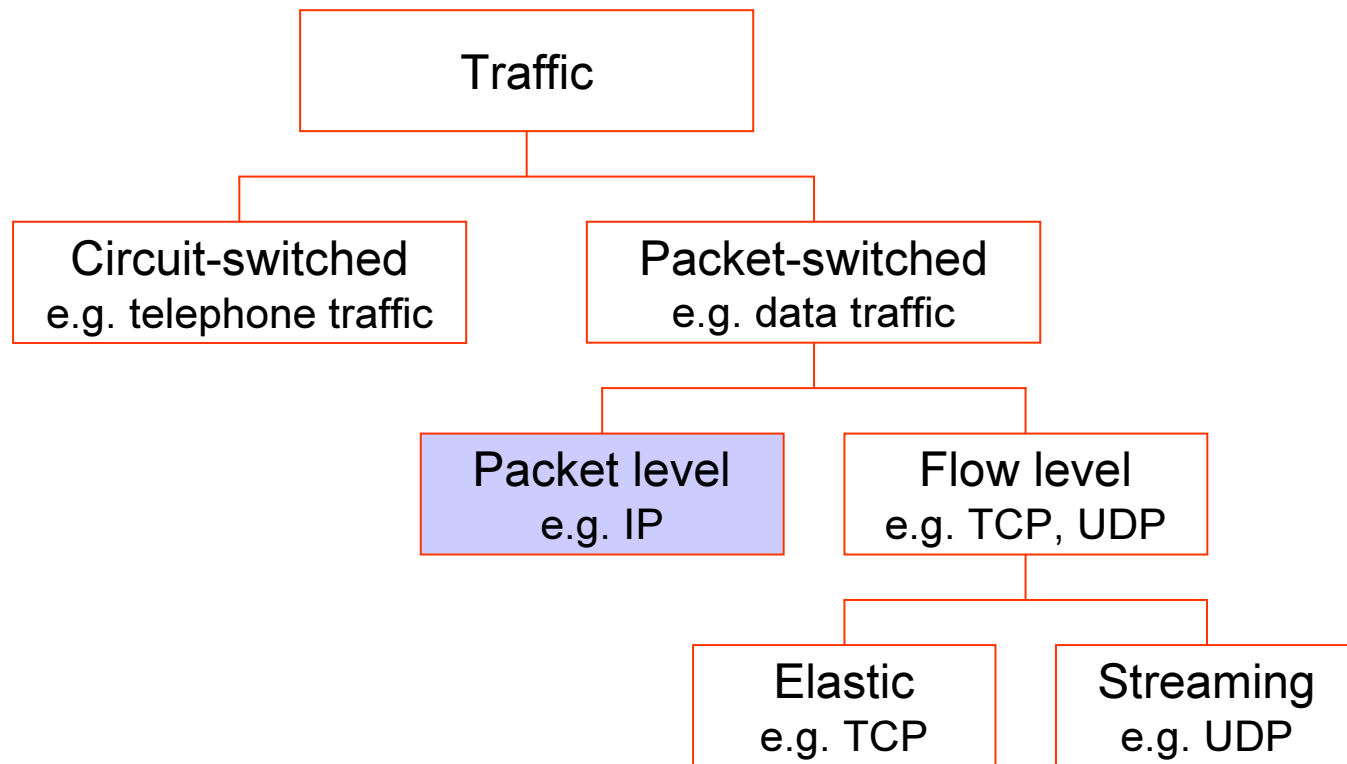


## Contents

- Traffic characterisation
- Telephone traffic modelling
- Data traffic modelling at packet level
- Data traffic modelling at flow level

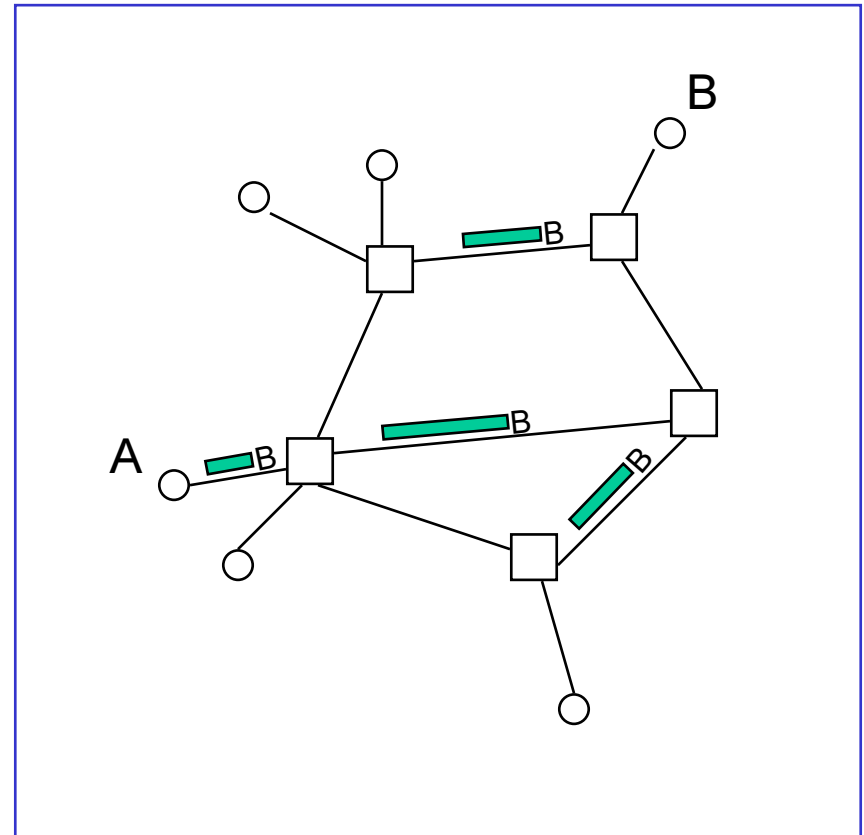


# Traffic classification



## Network layer in IP networks

- **IP** = Internet Protocol
- Connectionless:
  - no connection establishment
  - no resource reservations
- Information transfer as discrete packets
- **Best Effort** service paradigm
  - Network nodes (routers) forward packets “as well as possible”
  - Packets may be lost, delayed or their order may change
    - ⇒ “intelligence” should be implemented at the edge nodes or terminals

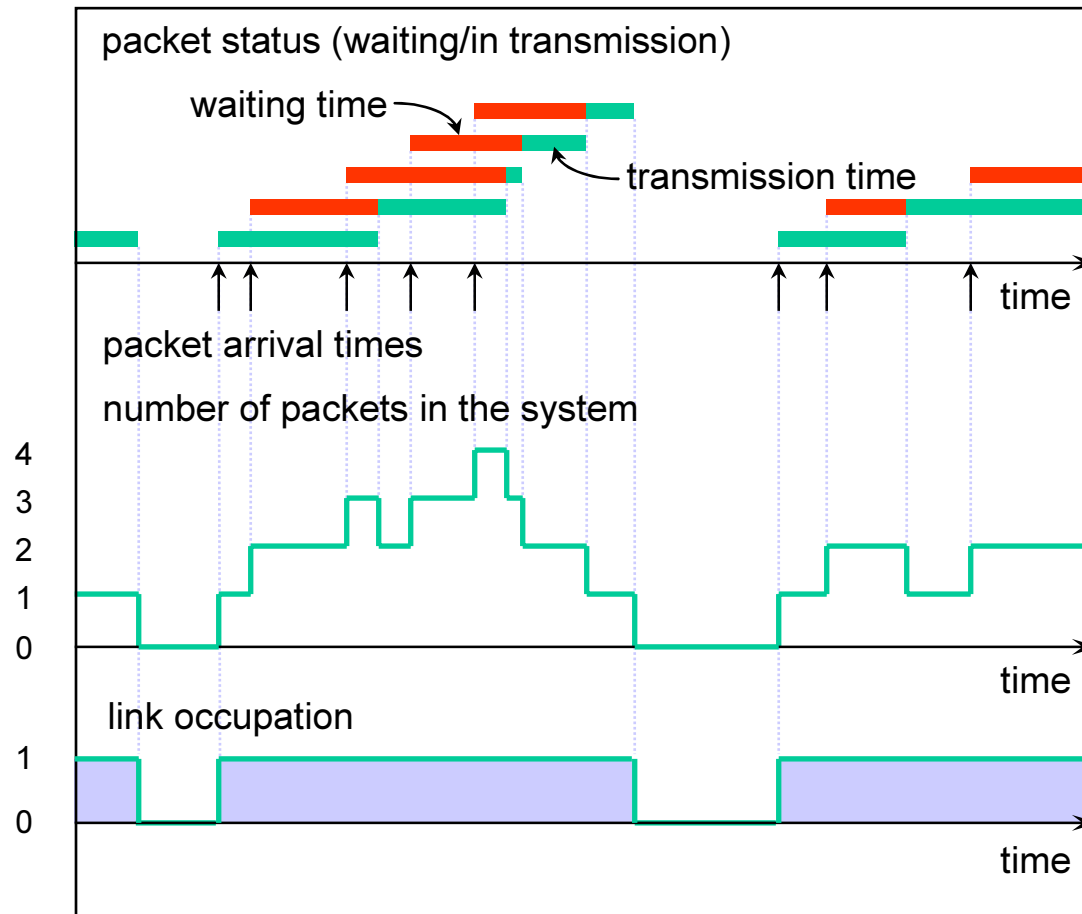


IP header	Data
-----------	------

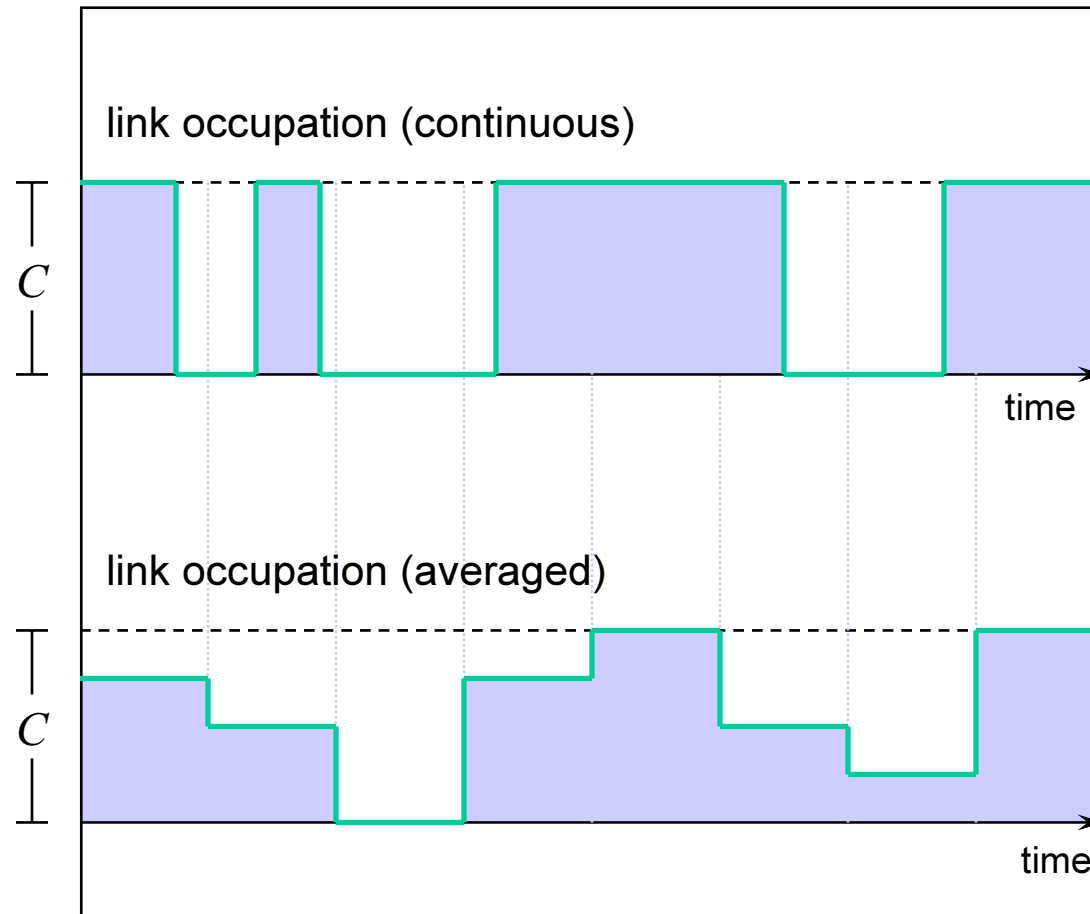
## Packet level model of data traffic

- Data traffic consists of **packets**
  - packets compete with each other for the processing and transmission resources (statistical multiplexing)
  - packet characterisation: **length** (in data units)
- Modelling of offered traffic:
  - **packet arrival process** (at which moments new packets arrive)
  - **packet length distribution** (how long they are)
- Link model: a **single server queueing system**
  - the service rate  $\mu$  depends on the **link capacity** and the **average packet length**
  - when the link is busy, new packets are buffered, if possible, otherwise they are lost
- Modelling of carried traffic:
  - **traffic process** tells the number of packets in the system (including both the packet in transmission and the packets waiting in the buffer)

## Packet level traffic process (1)



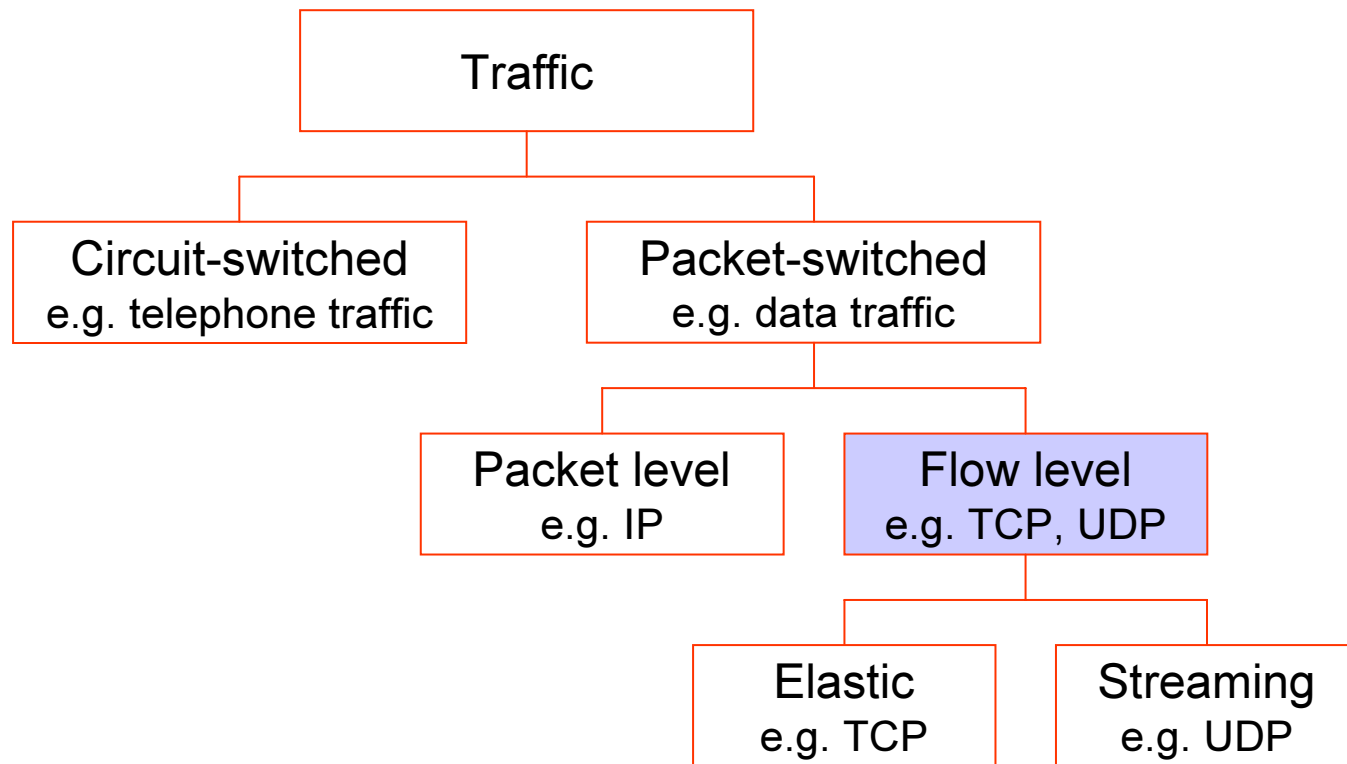
## Packet level traffic process (2)



## Contents

- Traffic characterisation
- Telephone traffic modelling
- Data traffic modelling at packet level
- Data traffic modelling at flow level

# Traffic classification



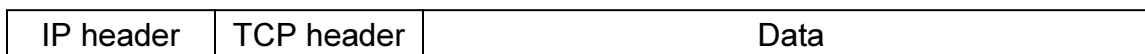
## Transport layer in IP networks

- On top of the network layer (IP) there is the **transport layer**
  - takes care of handling the IP packets in the terminals
  - operates **end-to-end**
- Transport layer protocols:
  - **TCP** = Transmission Control Protocol
    - transmission rate adapts to traffic conditions in the network by a congestion control mechanism
    - suitable for non-real time (elastic) traffic, such as transfers of digital documents
  - **UDP** = User Datagram Protocol
    - transmission rate independent of traffic conditions in the network
    - suitable for transactions (interactive traffic with short transfers)
    - used also for real time (streaming) traffic with the help of upper layer protocols, such as RTP



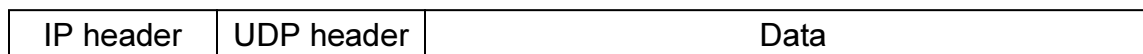
# TCP

- **TCP** = Transmission Control Protocol
  - **connection oriented** end-to-end transmission layer protocol
  - for a reliable byte stream transfer on top of IP
    - the delivery of packets in the right order is checked using acknowledgements and retransmissions
  - Protocol specific flow and congestion control mechanisms for traffic control
    - based on the use of an **adaptive sliding window**
  - **flow control**: prevents over flooding the receiver
    - the receiver tells how many bytes it can receive
  - **congestion control**: prevents over flooding the network
    - the transmitter has to find out when the network is congested
    - a **packet loss** indicates congestion: when a packet is lost, the window is decreased, otherwise gradually increased (to detect the network state)



# UDP

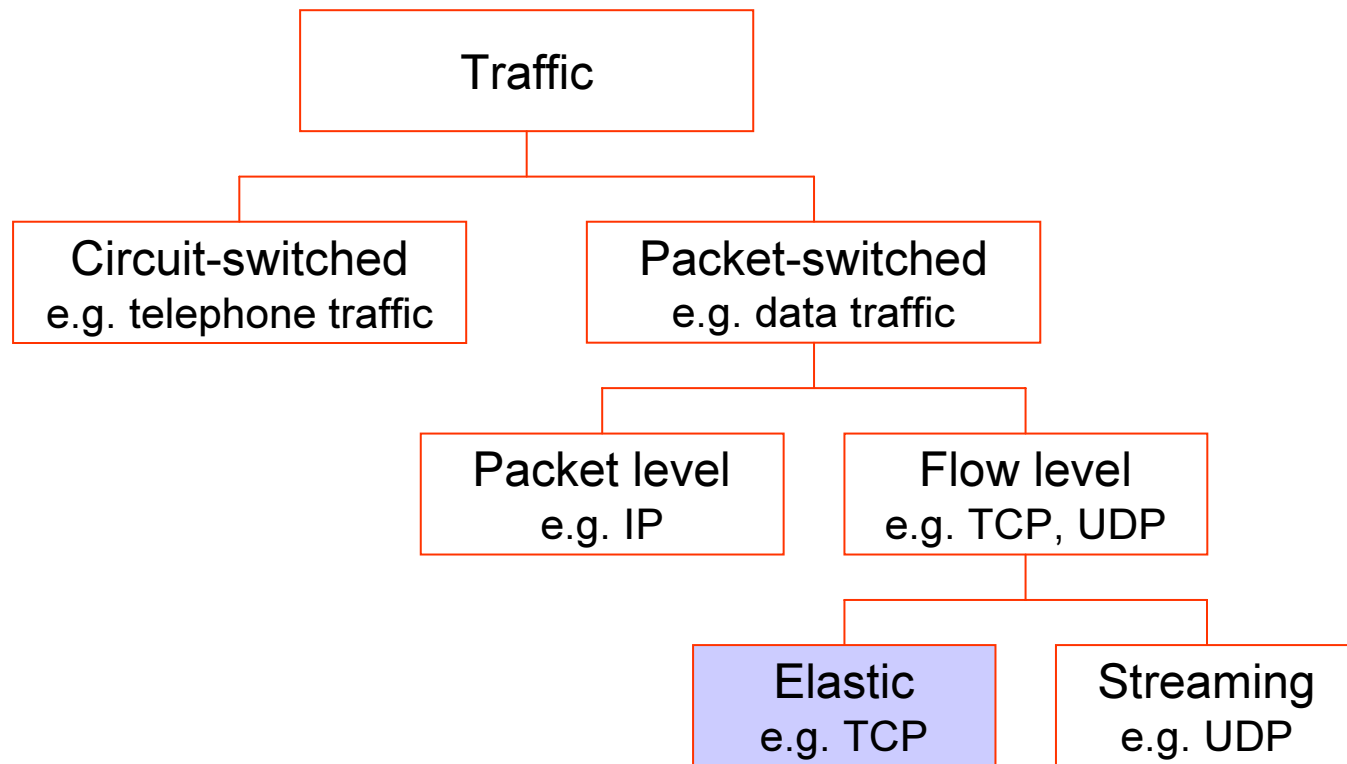
- **UDP = User Datagram Protocol**
  - **connectionless** end-to-end transmission layer protocol
  - on top of IP, but only for multiplexing
  - no guarantees of packet transfer (unreliable)
  - no flow control: may overload the receiver
  - no congestion control: may overload the network



## Data traffic at flow level

- In a longer time scale, data traffic may be thought to consist of **flows**
  - A single flow is described as a **continuous bit stream** with a possibly varying rate (and not as discrete packets)
- Flow classification:
  - **Elastic flows**
    - transmission rate adapts to traffic conditions in the network by a congestion control mechanism
    - e.g. transfers of digital documents (HTTP,FTP,...) using **TCP**
  - **Streaming flows**
    - transmission rate independent of traffic conditions in the network
    - e.g. real time voice, audio and video transmissions using **UDP**

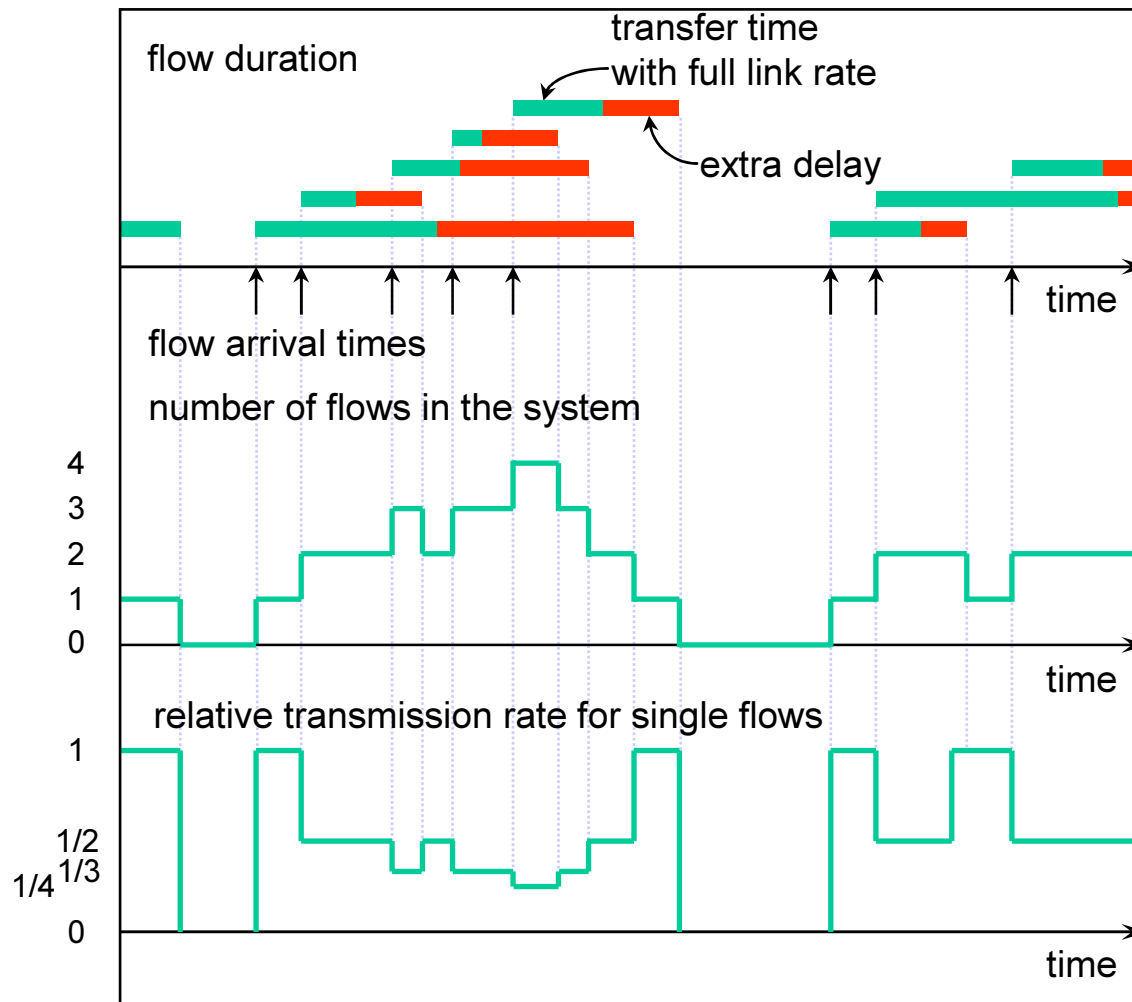
# Traffic classification



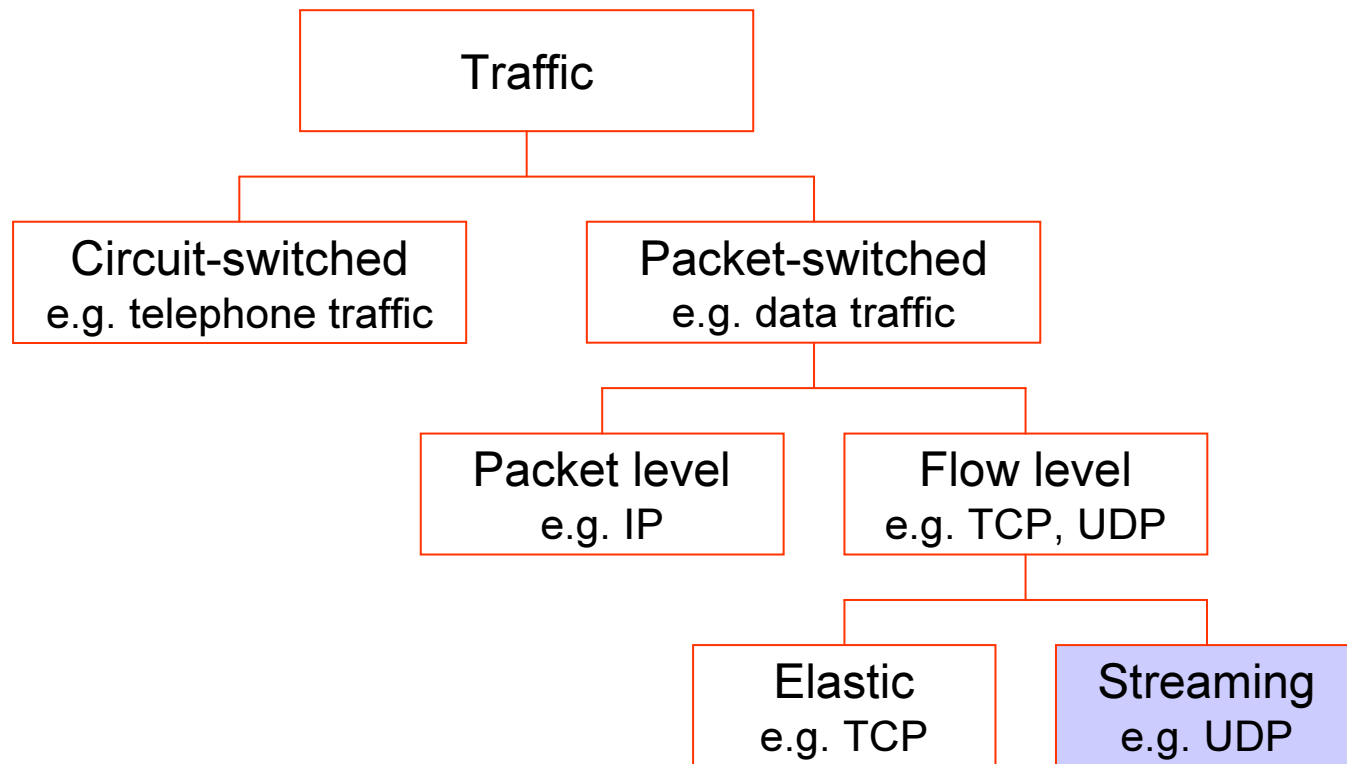
## Flow level model of elastic traffic

- Elastic traffic consists of adaptive TCP **flows**
  - flow characterisation: **size** (in data units)
  - the transfer rate and the duration of an elastic flow are not fixed but depend on the network state dynamically
- Modelling of offered traffic:
  - **flow arrival process** (at which moments new flows arrive)
  - **flow size distribution** (how large they are)
- Link model: a **sharing system**
  - due to lack of admission control, no flows are rejected
  - the service rate  $\mu$  depends on the **link capacity** and the **average flow size**
  - in the model, the adaptation of the transmission rate is immediate, and the link capacity is shared evenly (fairly) among all competing flows
- Modelling of carried traffic:
  - **traffic process** tells the number of flows in the system

## Flow level traffic process for elastic flows



# Traffic classification



## Streaming traffic classification

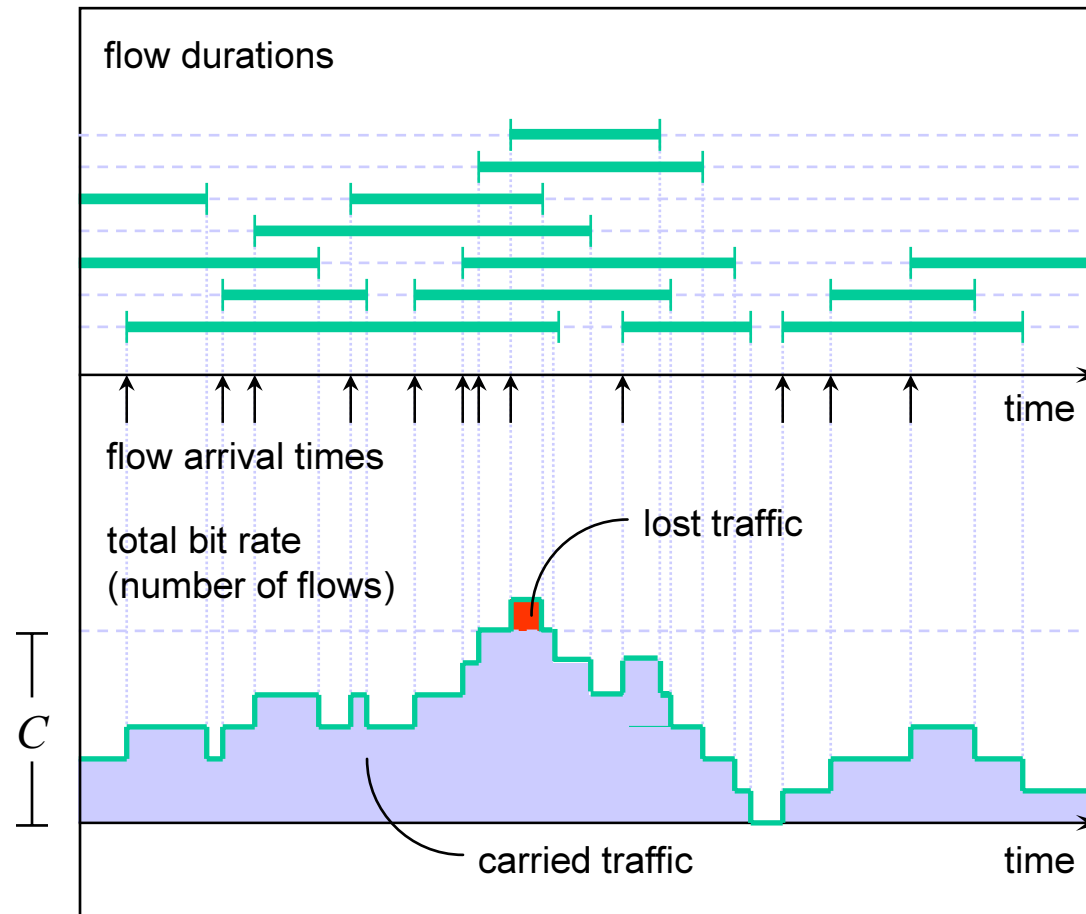
- **CBR** = constant bit rate
  - e.g. CBR coded voice/audio/video
  - packet level: fixed size packets generated regularly with uniform intervals
  - flow level: constant rate bit stream
  - flow characterisation: **bit rate** and **duration**
- **VBR** = variable bit rate
  - esim: VBR coded voice/audio/video
  - packet level: variable size packets generated irregularly
  - flow level: variable rate bit stream
  - flow characterisation: **bit rate as a function of time**



## Flow level model of streaming CBR traffic

- Streaming CBR traffic consists of UDP flows with constant bit rate
  - flow characterisation: **bit rate** and **duration**
- Modelling of offered traffic:
  - **flow arrival process** (at which moments new flows arrive)
  - **flow duration distribution** (how long they last)
- Link model: an **infinite system**
  - due to lack of admission control, no flows are rejected
  - the service rate  $\mu$  depends on the **average flow duration**
  - transmission rate and flow duration are insensitive to the network state
  - no buffering in the flow level model: when the total transmission rate of the flows exceeds the link capacity, bits are lost (uniformly from all flows)
- Modelling of carried traffic:
  - **traffic process** tells the number of flows in the system, and, as well, the total bit rate

## Flow level traffic process for streaming CBR flows



**THE END**

