

Institut für Rundfunktechnik



Video transport over IP – Can we still avoid the errors?

EBU Seminar:
Networks 2007
To IP and Beyond
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Video transport over IP

- Introduction
- MPEG-TS Encapsulation
- Protocol Overhead
- UDP
- RTP

Features of von IP- Networks

- IP-Network/Internet
- Overload
- Consequence for Real-time applications
- CoP #3
- Quality of Service
- Feature of CoS
- Experience: Managed Networks
- First Measurements

Conclusion

Video transport over IP



On this subject:

- contribution purpose
- high data rates
- Production quality
- post processing

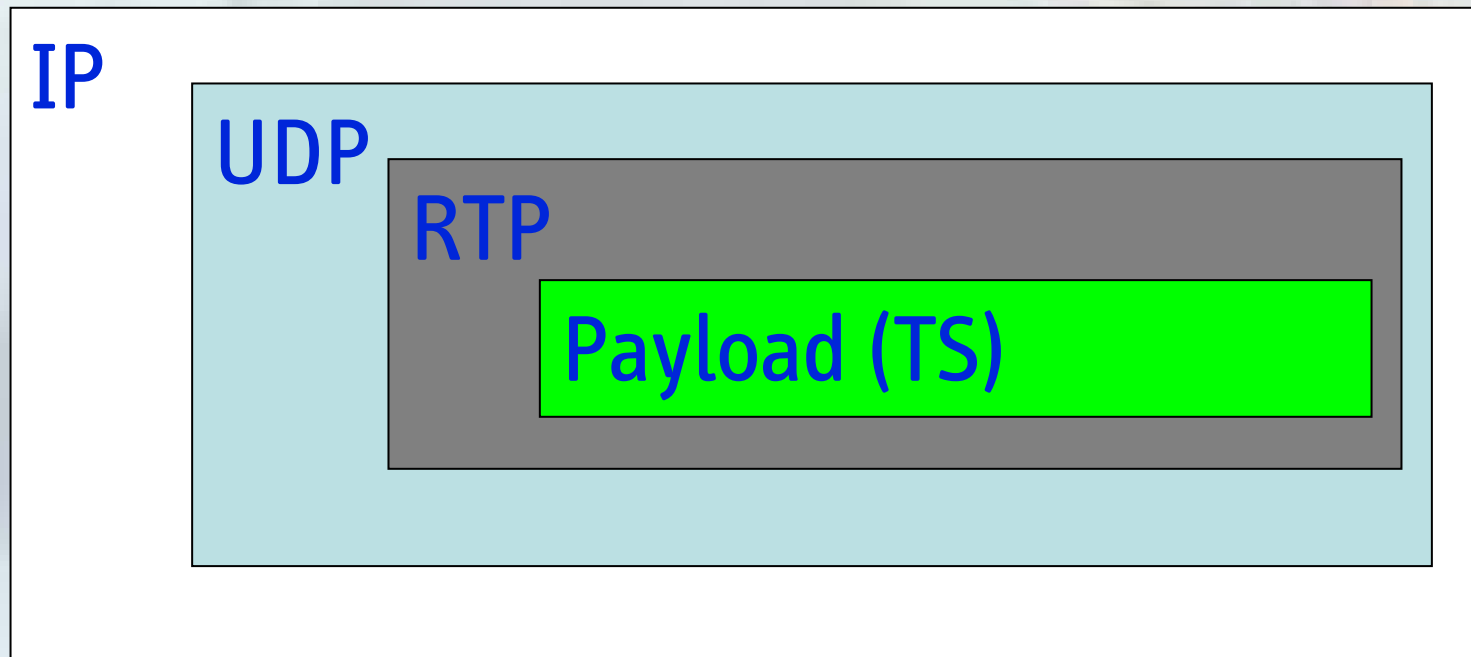
Not:

distribution propose
IPTV

MPEG-TS Encapsulation



Structure of an IP-Packet:



IP: Internet Protocol

UDP: User Datagram Protocol

RTP: Real-Time Transport Protocol

Protocol Overhead



	Header	Payload
Ethernet	18 Byte (22 VLAN)	46 – 1500 Byte
IP	20 Byte	26 -1480/(65.515) Byte
UDP	8 Byte	18 – 1472 Byte
RTP	12 Byte	6 – 1460 Byte
Overhead:	40 Byte per IP-Packet	Up to 7 MPEG-TS Packets (188 Byte, Ethernet)

Bit error and UDP



UDP-Header:

Source Port
Destination Port
Length
Checksum

Checksum:

Verification: if bit errors happen
error detection – no error correction possible!
in case → reject IP-Packet!!!

A single bit error always causes the loss of the complete IP-Packet!

Main features of RTP:

Sequence numbering:

- Detect packet loss
- Detect out of order packets

Timestamp

- Synchronisation purpose
- Calculate network jitter regardless PCR

PCR: Primary Clock Reference

Characteristics of IP-Networks



IP-Network/Internet

Overload

Consequence for Real-time applications

CoP #3

Quality of Service

Feature of CoS

Experience: Managed Networks

Internet - IP Networks



IP-Networks \neq Internet

Internet: based on Internet Protocol

No QoS

Everything can happen at every time

IP-Network:

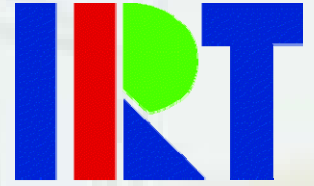
Managed network

Defined QoS

Deterministic behaviour (hopefully)

every single IP-Packet competes with each other

- intermediate storage within routers (individual delay, jitter)
- buffer over flow
- rejected packets
- every connections and data streams equally concerned!
- no difference witch contents is carried!



Consequence for Real-time applications

bit error (UDP)	→ packet loss
overload	→ packet loss
delay (Jitter)	→ packet loss

→ Consequence: forward error correction?

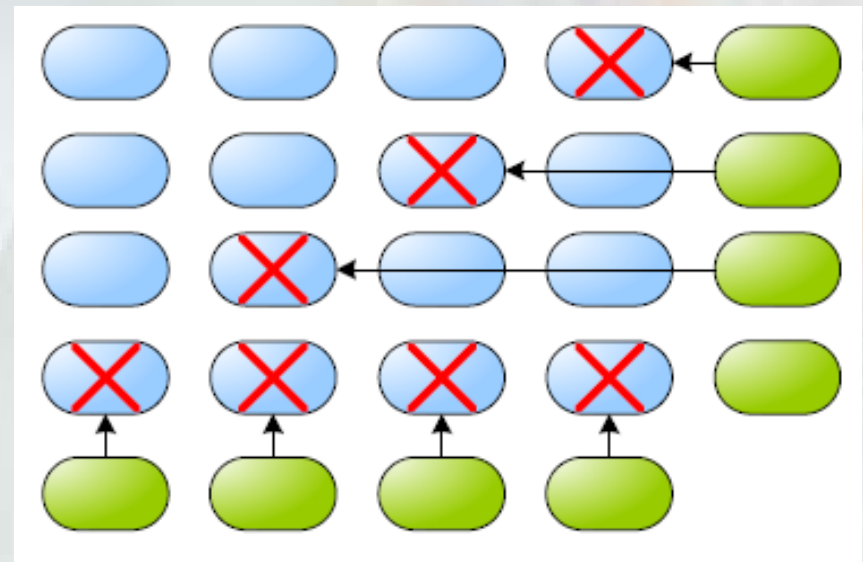
Pro MPEG Forum: CoP #3



Codes of Practice #3 : Forward Error Correction (FEC) for MPEG-TS

Matrix: L columns, D rows
Correction of burst errors
possible!

Long delay
(particularly at low data rates)



MPEG: Moving Picture Experts Group

Feature: CoP #3



Data stream, Column FEC and row FEC are separate streams

FEC Overhead: 5 – 30 % redundant data necessary!

FEC-Packets are interleaved with data packets

→ to avoid large traffic rate changes (traffic shaping)

→ additional delay: depend on the arrangement of packets (up to $L \times D$ packet time)

Quality of Service (QoS)



Priority of Packets (DiffServ)

- 3 Bit for classification → 7 Class of service (CoS)
- Typical 4 for users
- 2 classes for OAM

5 VoIP

Voice over IP, Audio

4 Multimedia

e.g. Videoconferences, Video over IP

3 Privilege

Preferred data applications

1 Best Effort

all others (Email, www, ...)

QoS: Quality of Service

CoS: Class of Service

OAM: Operation, Administration and Maintenance

VoIP: Voice over Internet Protocol

DiffServ: Differentiated Services

QoS Parameter



QoS Class	Packet loss	Round Trip Delay Backbone (domestic)
VoIP	0,10%	30 ms
Multimedia	0,20%	35 ms
Privilege	0,30%	40 ms
Best Effort	1 %	60 ms

Feature of service class



→ every connection and data within a service class (CoS) have equal rights!

Overload within a service class:

→ Reject of packets

→ **every** connection is concerned!!!

Example (multimedia class has a capacity of 10 video streams):

A additional video (11th) connection release overload

→ Rejection of packets

→ **every** present video connection is concerned!!!

→ additional mechanism necessary!

CoS: Class of Service

Experience: Managed Networks



Bit error: → single packet loss

Correctable by FEC

Overload: → periodic packet loss

easy to correct by FEC (within limits, low loss rate)

Re-order: → atypical for managed networks

Except auto configuration is used

Switch over (equivalent network, cut-off, auto configuration):

→ burst errors

burst rate depend on data rate

common not compensation able

Interrupt!

First Measurements



Configurations:

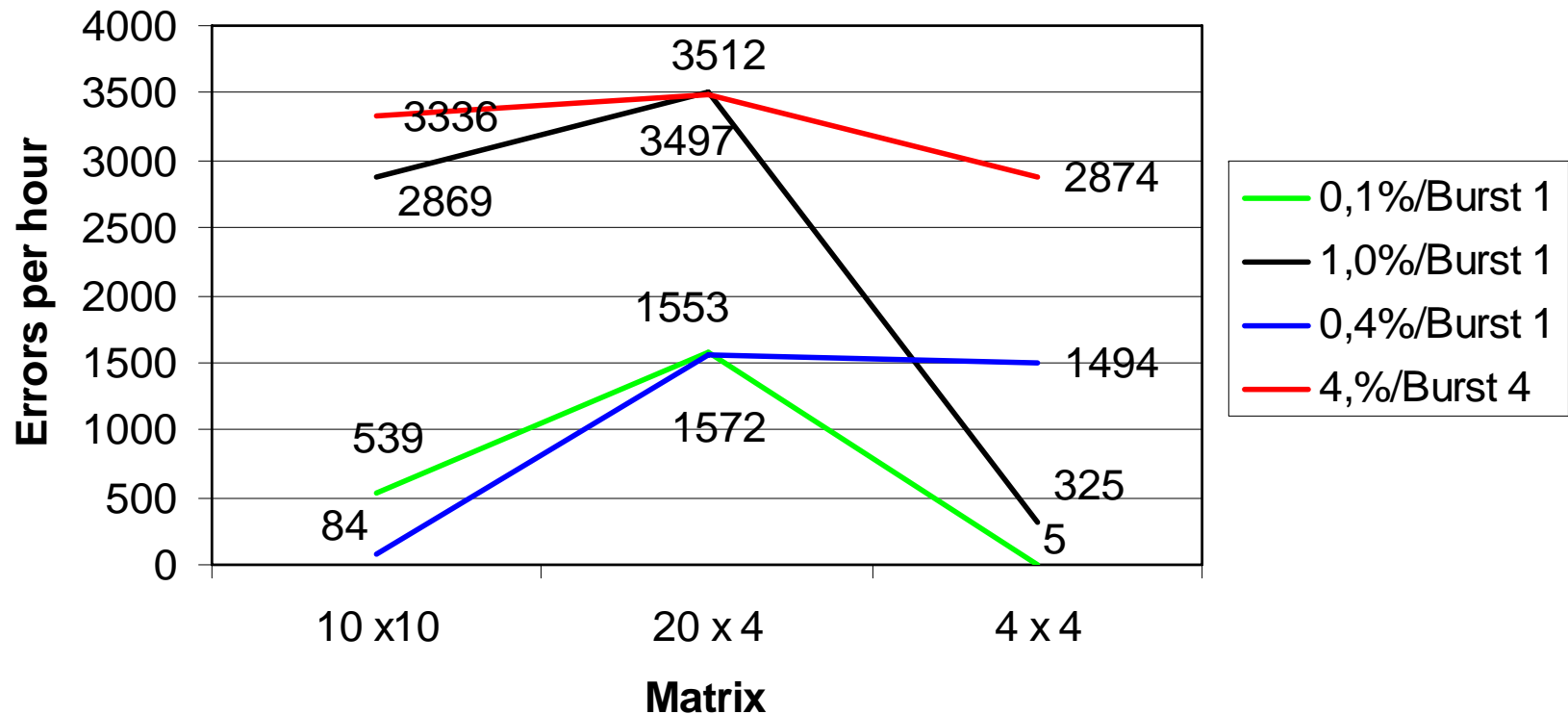
video rate: 10 Mbit/s

Matrix (CoP 3)	Max. corrigible errors	Delay
4x4	4	25 ms
10 x 10	10	180 ms
20 x 4	4	142 ms

First Measurement Results (1)



Pro-MPEG Forum FEC / Random Errors



First Measurement Results (2)



Random Errors:

Data and corresponding FEC packets are lost

Constellations with the FEC cannot correct always exist

Periodic Errors:

Every lost packet is corrected

Behaviour of low error rates

Conclusion



Managed networks increase deterministic behaviours
But IP-Networks are not ready for real-time broadcast
quality!!!

Nevertheless additional mechanism is necessary to prevent
packet loss (e.g. FEC)

IP tax: FEC increase bandwidth and delay

Further improvement necessary

→ no guarantee for real-time video and audio!

**Thank you
for your attention !**



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