

Efficient QoS Support for Voice-over-IP Applications Using Selective Packet Marking

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Overview

- Voice over IP (VoIP)
 - Improved quality for VoIP
- Approach
 - Performance of the G.729 loss concealment
 - Speech Property-Based Selective Packet Marking
- Evaluation
 - Reference packet marking/prioritization schemes
 - Simple network models
 - Objective speech quality measurement
- Conclusions

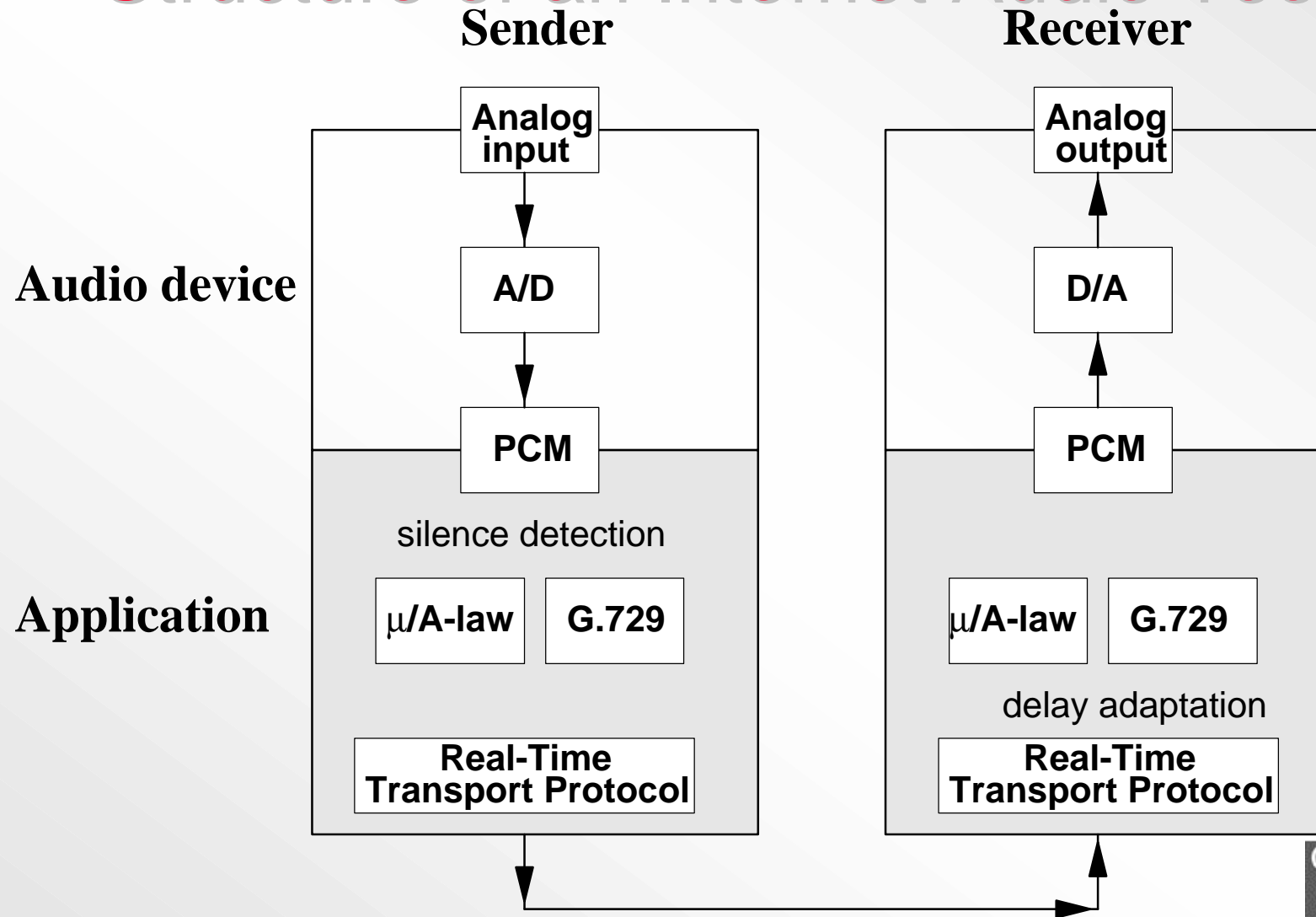
Voice over IP

- Main drivers:
 - current economical incentives (Internet flat rate pricing) → *Internet Telephony*
 - service integration, unified packet-switching infrastructure
- One of the main problems:
 - satisfaction of real-time QoS demands in a packet-switched network (fundamental tradeoff: statistical multiplexing vs. reliability → *packet loss*)

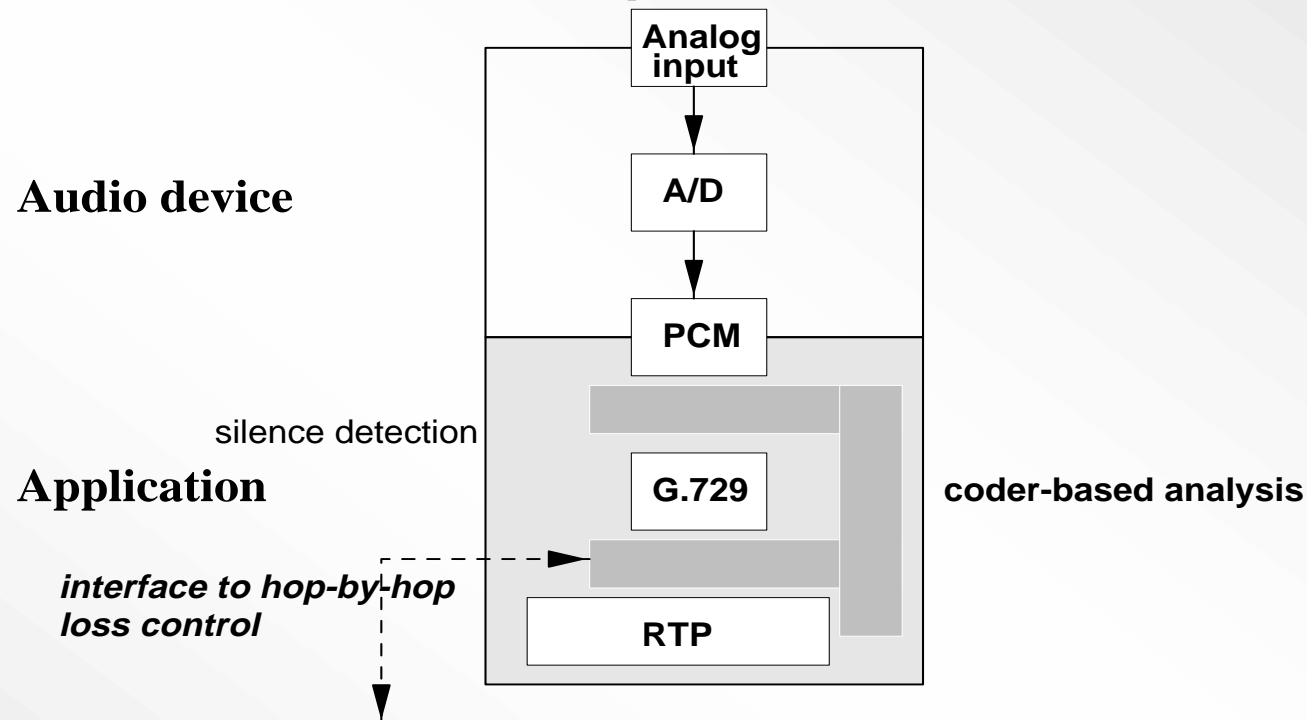
QoS for Voice over IP flows

- low bitrate → high per-flow overhead for reservations (scalability, need for multiplexing)
- high compression (backward adaptive coding: ITU-T G.729, G.723.1)
 - no further sender adaptation / network adaptation (transcoding) possible
 - amplifies high perceptual impact of burst losses (error propagation)
- + tolerance to isolated losses (speech stationarity → extrapolation of coder state → loss concealment) → extend the loss resiliency of high-compressing codecs using prioritization (priority enforcement on flow aggregates: Differentiated Services)

Structure of an Internet Audio Tool

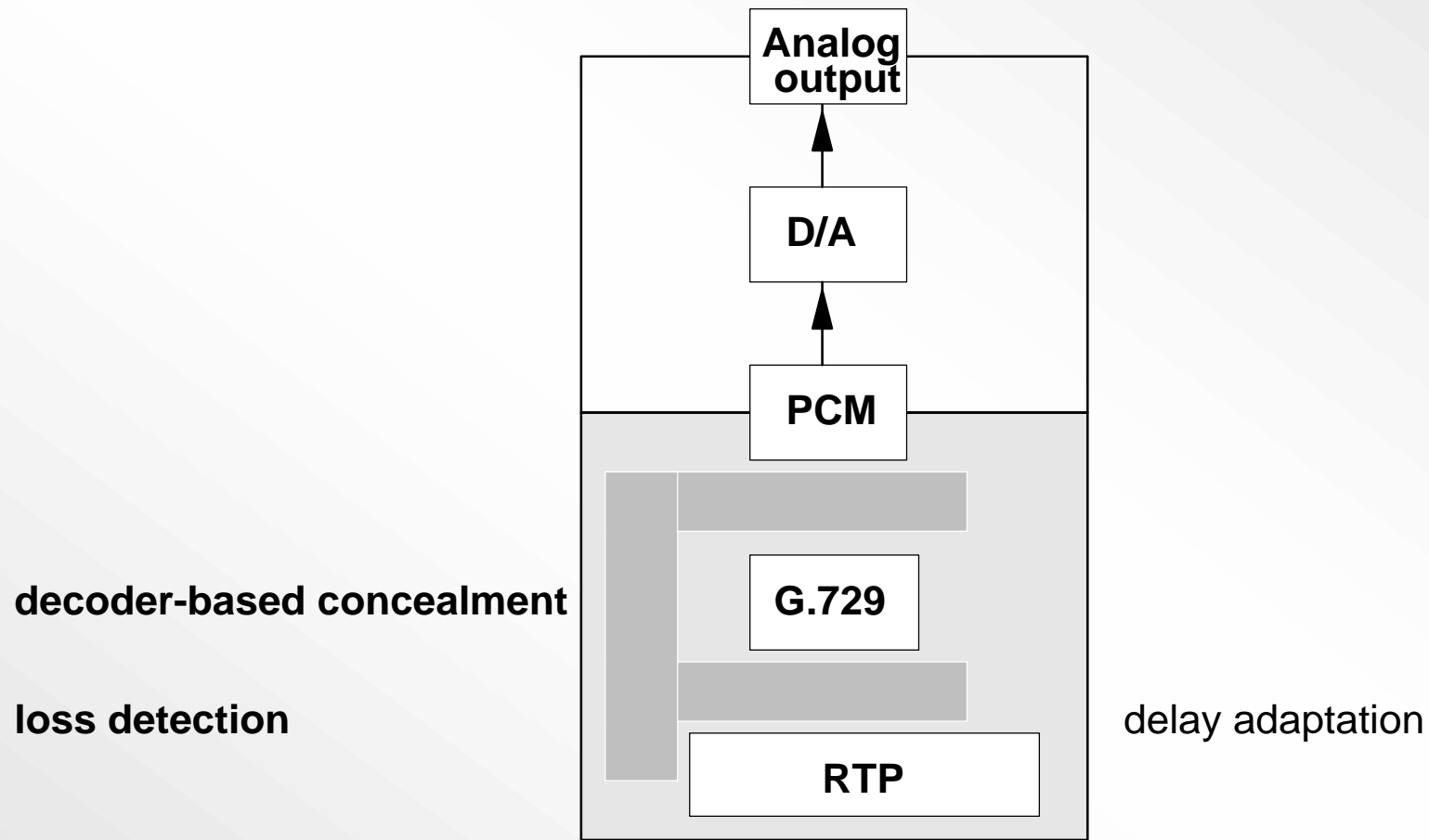


Additional components: Sender



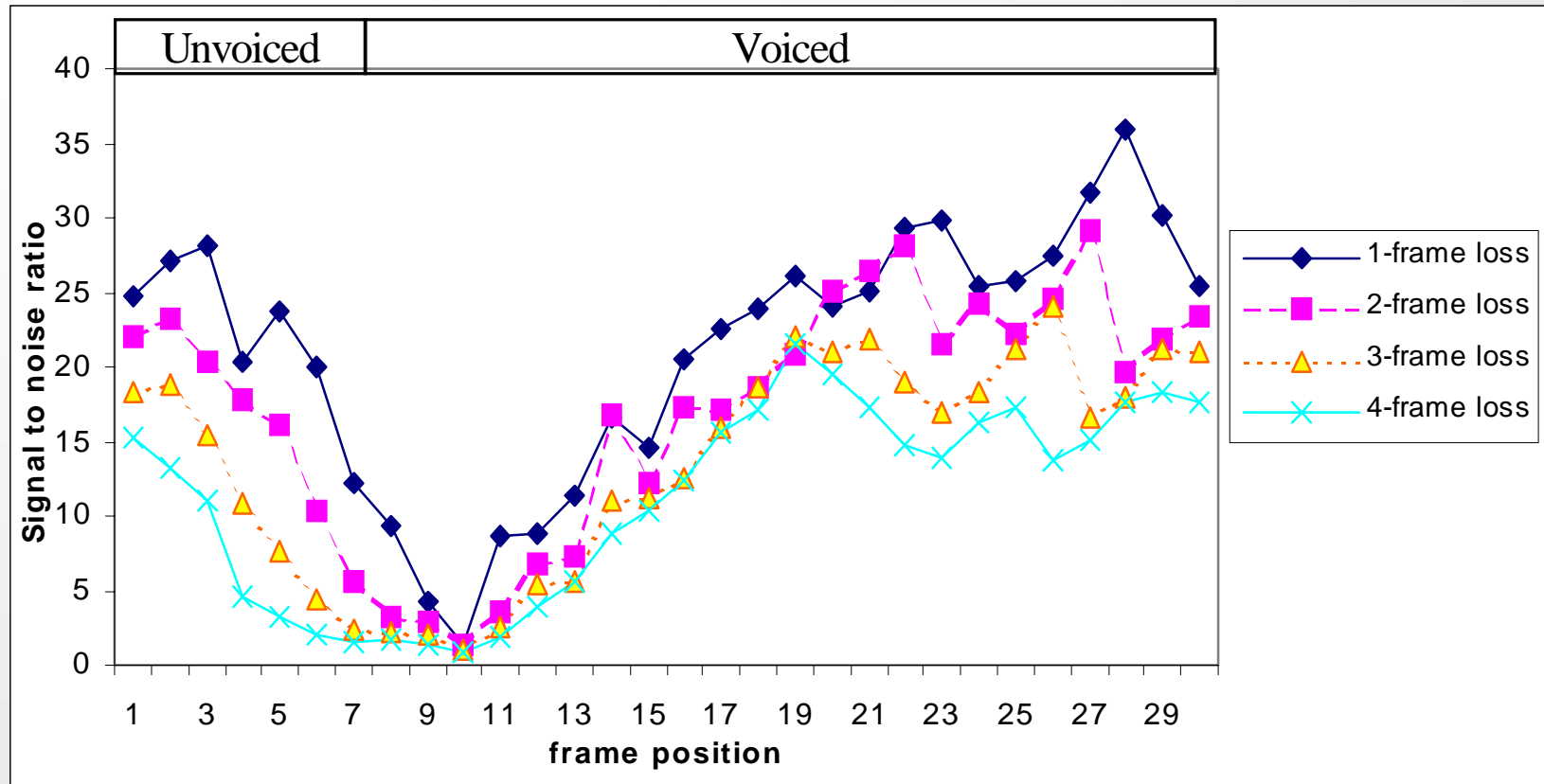
- Side information available at the encoder is used
 - Decoder concealment process is taken into account
- Selective packet marking

Receiver



- No generic (PCM-level) concealment

Performance of G.729 loss concealment



- Decoder fails to conceal losses at unvoiced/voiced transition due to lack of state (synthesis filter parameters, excitation)

Speech Property-Based Selective Packet Marking

```
protect = 0
foreach (k frames)
  classify = analysis(k frames)
  if (protect > 0)
    if (classify == unvoiced)
      protect = 0
      send(k frames, "0")
    else
      send(k frames, "+1")
      protect = protect-k
    endif
  else
    if (classify == uv_transition)
      send(k frames, "+1")
      protect = N-k
    else
      send(k frames, "0")
    endif
  endif
endif
endfor
```

- Packet marking („0“, „+1“) is application-controlled (adaptive to expected loss concealment performance): *40-50% of packets marked with higher priority („+1“)*

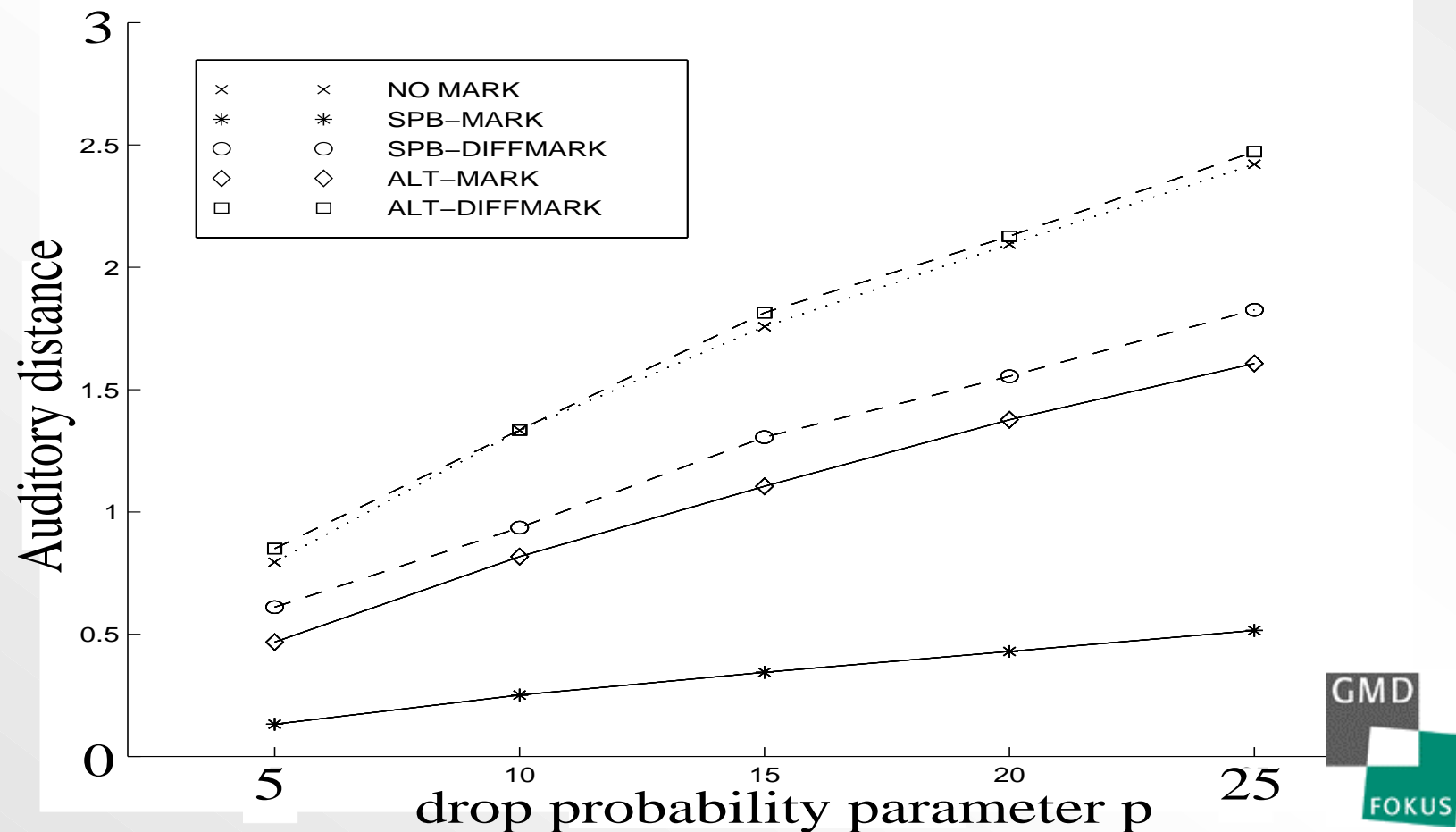
Marking Scheme

/ Network Model

NO MARK	<table border="1"> <tr> <td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td> </tr> </table>	0	0	0	0	0	0	
0	0	0	0	0	0			
FULL MARK	<table border="1"> <tr> <td>+1</td><td>+1</td><td>+1</td><td>+1</td><td>+1</td><td>+1</td> </tr> </table>	+1	+1	+1	+1	+1	+1	
+1	+1	+1	+1	+1	+1			
SPB MARK	<table border="1"> <tr> <td>0</td><td>+1</td><td>+1</td><td>0</td><td>0</td><td>0</td> </tr> </table>	0	+1	+1	0	0	0	
0	+1	+1	0	0	0			
ALT MARK	<table border="1"> <tr> <td>0</td><td>+1</td><td>0</td><td>+1</td><td>0</td><td>+1</td> </tr> </table>	0	+1	0	+1	0	+1	
0	+1	0	+1	0	+1			
SPB DIFFMARK	<table border="1"> <tr> <td>0</td><td>+1</td><td>+1</td><td>-1</td><td>-1</td><td>0</td> </tr> </table>	0	+1	+1	-1	-1	0	
0	+1	+1	-1	-1	0			
ALT DIFFMARK	<table border="1"> <tr> <td>-1</td><td>+1</td><td>-1</td><td>+1</td><td>-1</td><td>+1</td> </tr> </table>	-1	+1	-1	+1	-1	+1	
-1	+1	-1	+1	-1	+1			

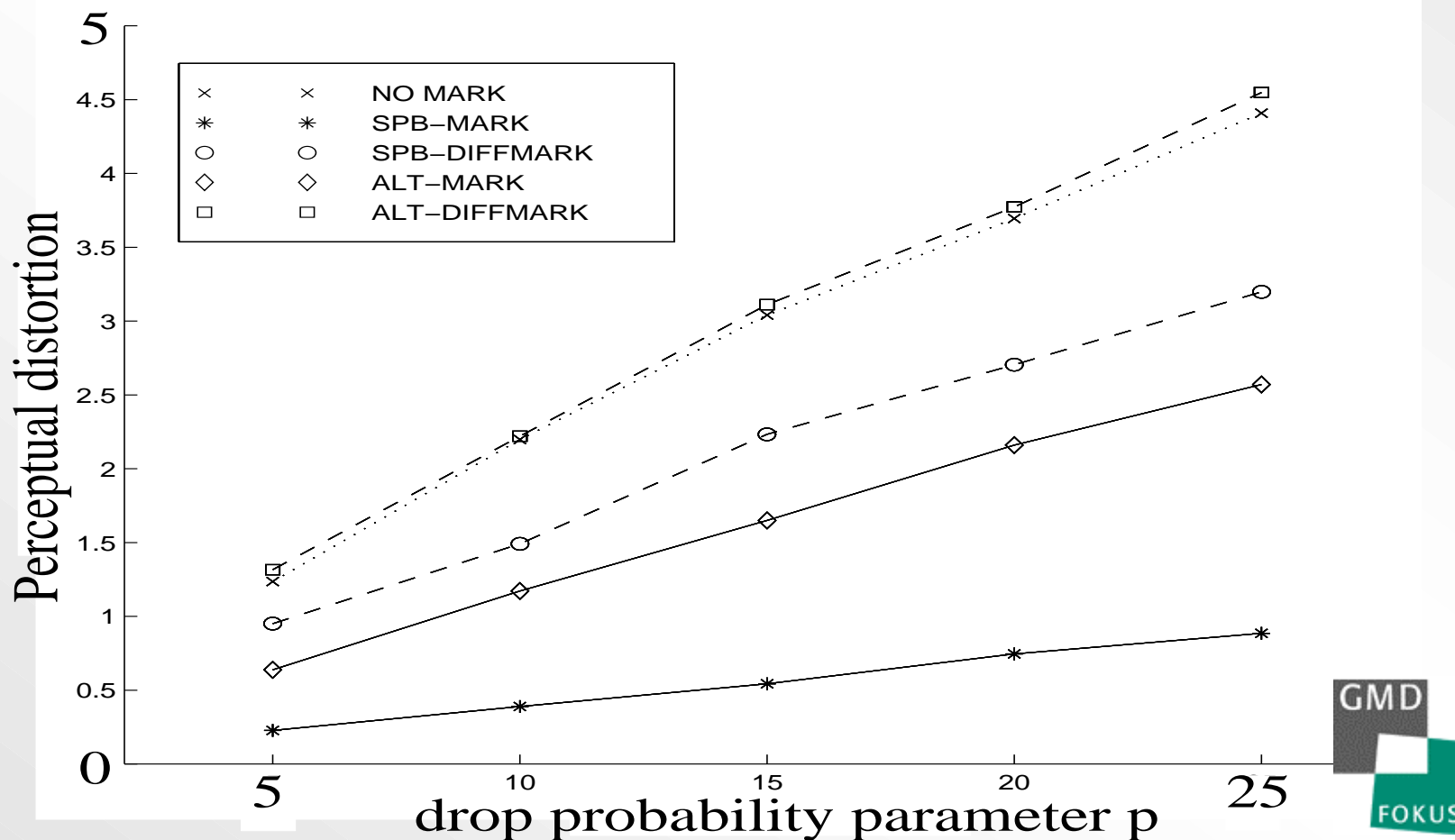
Results: Auditory Distance

- Application of recent advances in objective speech quality measurement: ITU P.861A



Results: Perceptual Distortion

- Enhanced Modified Bark Spectral Distortion (EMBSD; Temple University)



Conclusions

- SPB-(DIFF)MARK exploits differences in „concealability“ to adjust the amount of needed network prioritization
- simple network models & objective speech quality measures showed the reduction of number of marked packets while maintaining a good output quality
- selective marking requires appropriate socket interface (or application-level gateway)
- future work: employ speech quality measurement for the marking decision
- <http://www.fokus.gmd.de/g1one>

