

MPEG-1

Overview of MPEG-1 Standard

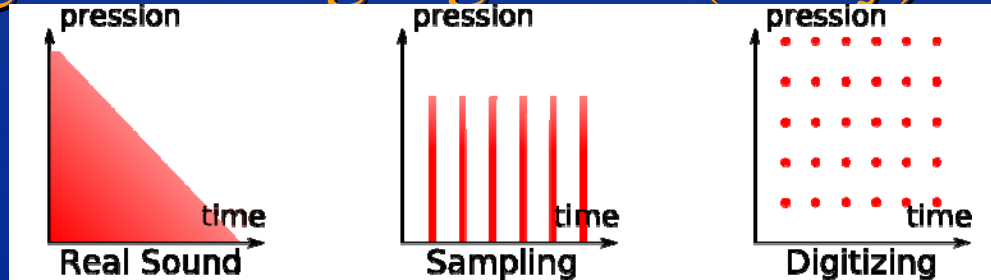
Introduction to perceptual and entropy codings

Contents

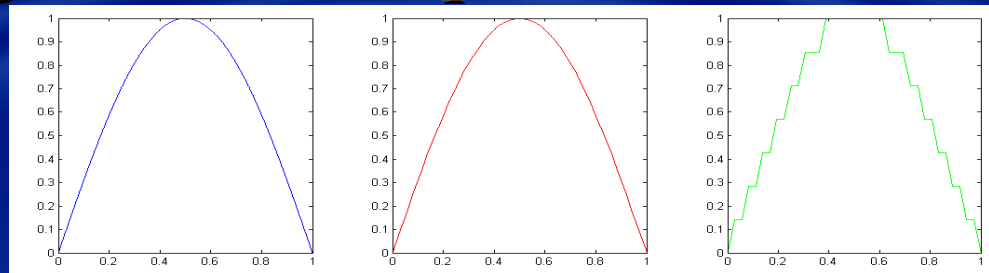
- History
- Psychoacoustics and perceptual coding
- Entropy coding
- MPEG-1
 - Layer I/II
 - Layer III (MP3)
 - Comparison and Audio Quality

Introduction

- Digitizing an analog signal is (lossy) compression



- Digitizing introduces quantization noise



- Quantization noise imply loss of quality

- Linear quantization > 16 bit (98 dB) \rightarrow inaudible noise (CD)
- Linear quantization 4 bit (26 dB)



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History

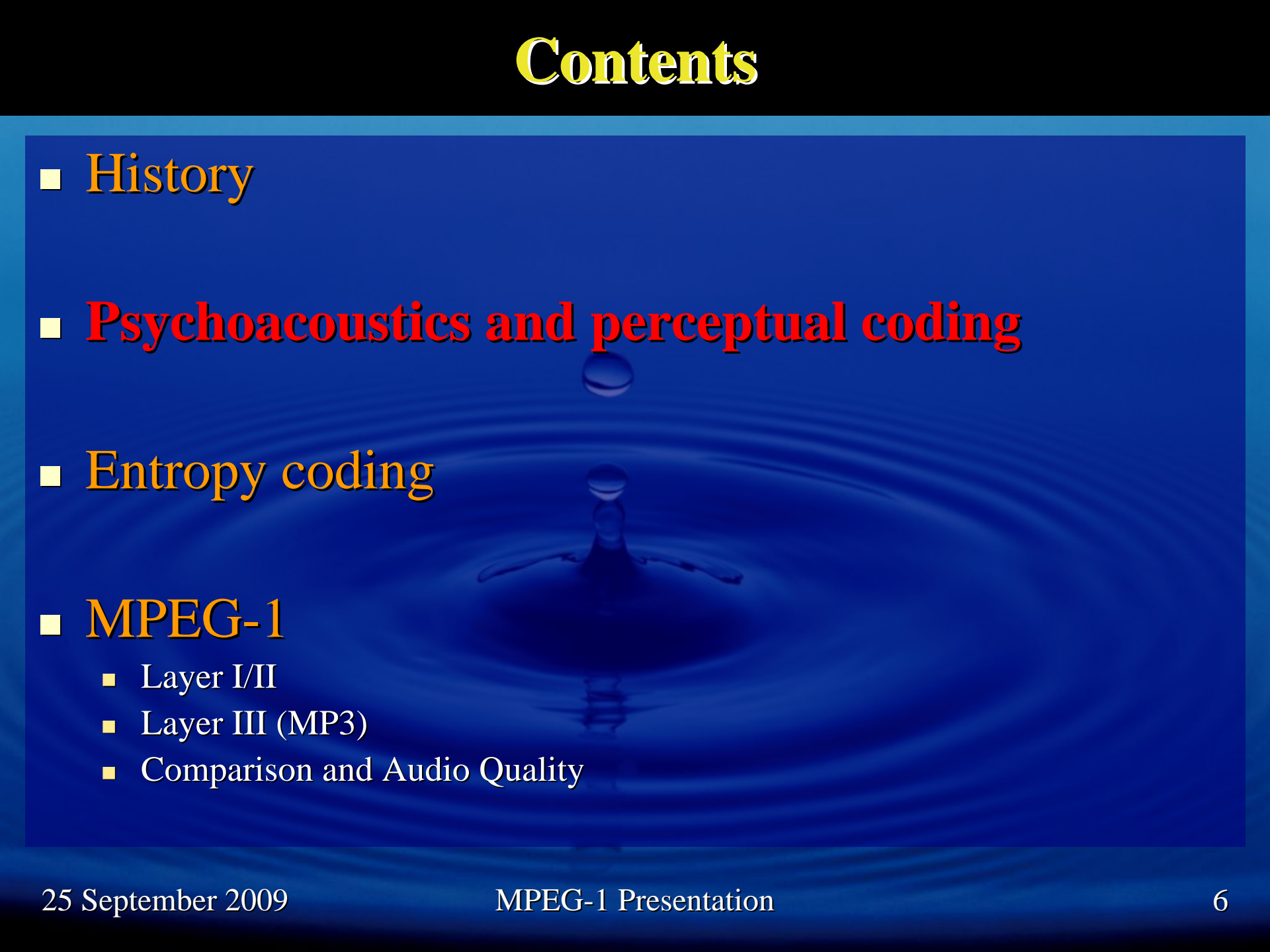
■ Moving Picture Expert Group (MPEG)

- Created in January 1988
- Starts the development of MPEG-1 in May 1988
- Publishes the MPEG-1 standard in November 1992 (ISO/IEC 11172-3 for audio)

■ MPEG-1 standard

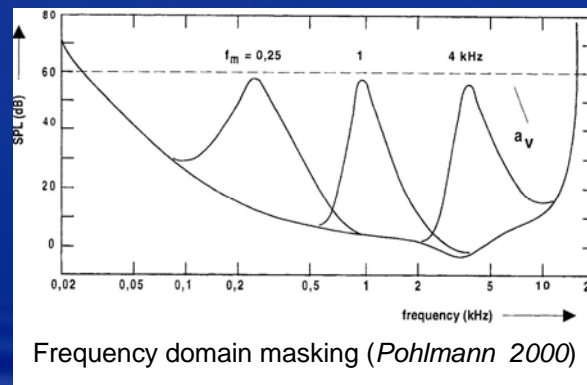
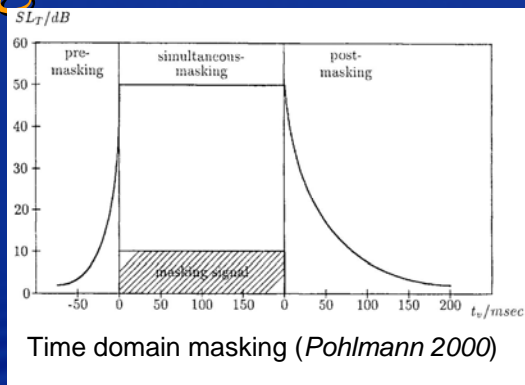
- Defines bit-stream
- Defines decoding functions
- DOES NOT define encoding techniques
- Inspired by MUSICAM (Masking pattern Universal Subband Integrated Coding And Multiplexing)

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Psychoacoustics

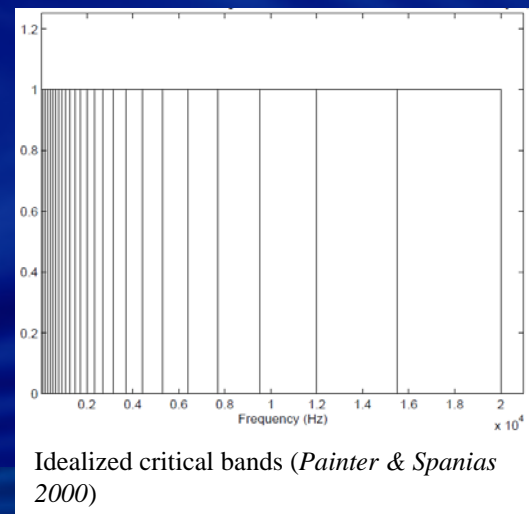
■ Masking effect



Critical bands

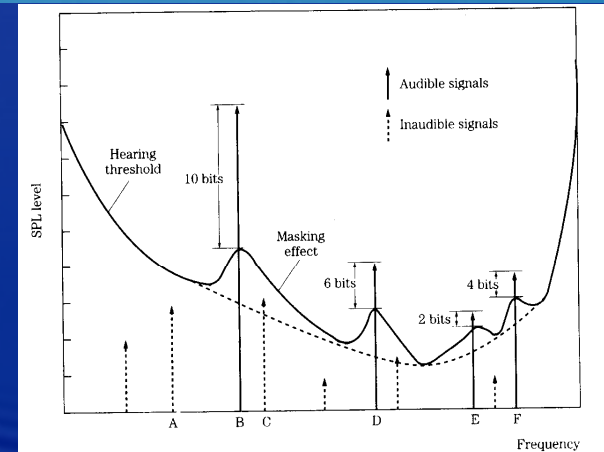
z/Bark	lower boundary	higher boundary	bandwidth	central frequency
0	0	100	100	50
1	100	200	100	150
2	200	300	100	250
3	300	400	100	350
4	400	510	110	450
5	510	630	120	570
6	630	770	140	700
7	770	920	150	840

(Brandenburg)

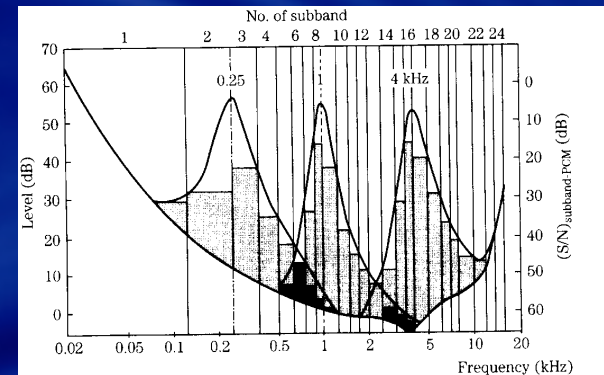


Perceptual Coding

- Dividing the different subbands of a signal
- Ignoring masked audio information
- Introducing inaudible quantization noise

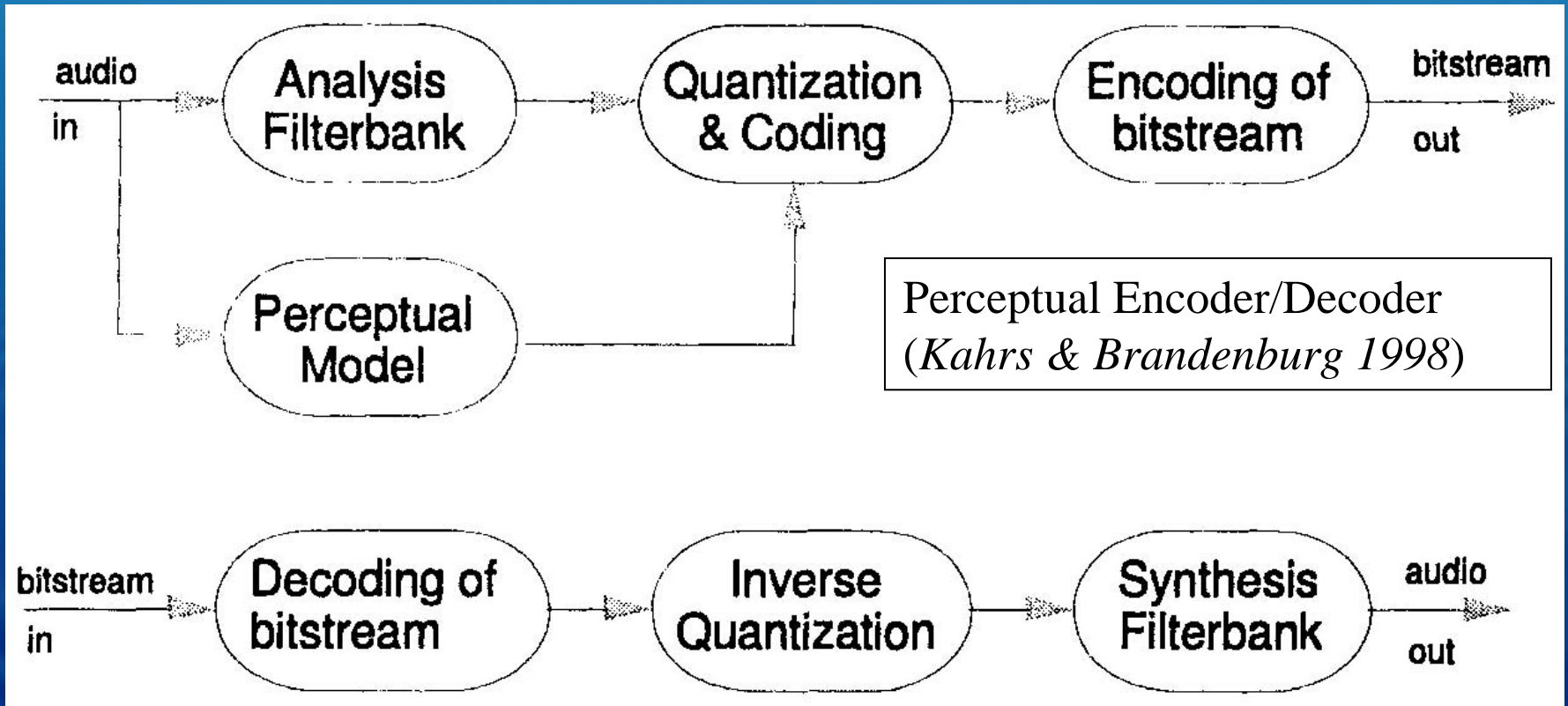


Bits association according to masking threshold (*Pohlmann 2000*)



Quantization noise added according to masking threshold (*Pohlmann 2000*)

Perceptual Coding



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Entropic Coding

- Use information about the signal to code efficiently
- Entropy of a signal
 - Example 1: {0, 2, 2, 2, 0, 0, 0, 0, 0, 2, 0, 3, 2, 2, 0, 0, 0, 3, 0, 0}
20 symbols – twelve 0 (0.6), zero 1 (0), six 2 (0.3), two 3 (0.1)
Entropy $H = 1.30$
 - Example 2: {1, 2, 3, 0, 2, 1, 1, 2, 3, 0, 0, 1, 0, 3, 3, 3, 2, 0, 1, 2}
20 symbols – five 0 (0.25), five 1 (0.25), five 2 (0.25), five 3 (0.25)
Entropy $H = 2$
- Shannon theorem
 - It is impossible to code with less than H bits/symbol
 - It is possible to code with less than $H+1$ bits/symbol

Entropic Coding

■ Huffman coding

- Example 1: {0, 2, 2, 2, 0, 0, 0, 0, 0, 2, 0, 3, 2, 2, 0, 0, 0, 3, 0, 0}
- 20 symbols – twelve 0 (0.6), zero 1 (0), six 2 (0.3), two 3 (0.1)
- Entropy $H = 1.30$

- Immediate coding:

0 → “00” 1 → “01” 2 → “10” 3 → “11”

“0010101000000000001000111010000000110000”

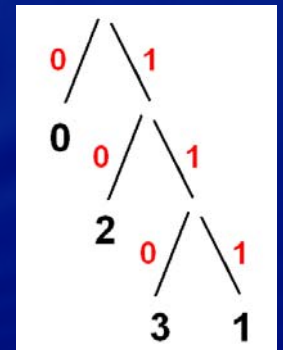
- Huffman coding:

0 → “0” 1 → “111” 2 → “10” 3 → “110”

“010101000000100110101000011000”

■ Efficiency:

- Immediate coding: 2 bits/symbol
- Huffman coding: 1.5 bits/symbol (statistically)



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MPEG-1

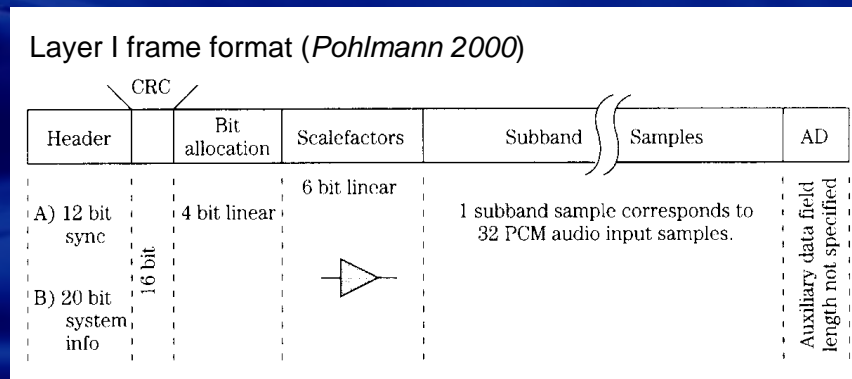
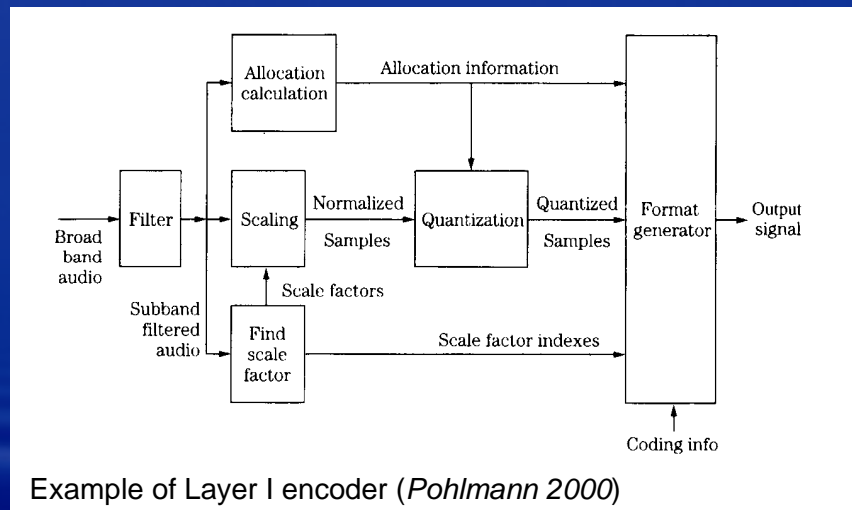
- Sampling rate: 32, 44.1 and 48 kHz
- Four modes:
 - Mono: 1 channel
 - Stereo: 2 channels
 - Dual: 2 channels independent (e.g. bilingual programmes)
 - Joint stereo: 2 channels coded together
- 2 perceptual models
- Floating point quantization (normalization)
- Error checking: Cyclic redundancy check (CRC)

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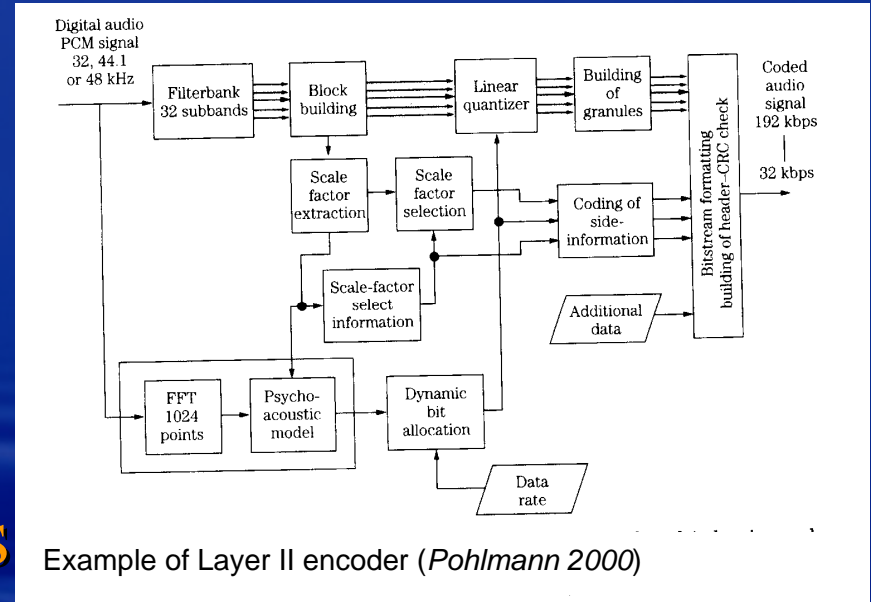
MPEG-1 Layer I

- From 32 to 448 kbps
 - 32-subband polyphase filterbank
 - Bit allocation (0-15)
 - Max dynamic range > 120 dB
 - Linear quantization
 - 1 frame \Leftrightarrow 384 samples
- Example: Philips Digital Compact Cassette



MPEG-1 Layer II

- From 32 to 384 kbps
- Improvement of Layer I
 - Improved FFT analysis
 - Scale factor redundancy
 - Finer quantization
- 1 frame \Leftrightarrow 1152 samples



Example of Layer II encoder (Pohlmann 2000)

Example: Digital Audio Broadcasting (DAB)

Layer II frame format (Pohlmann 2000)

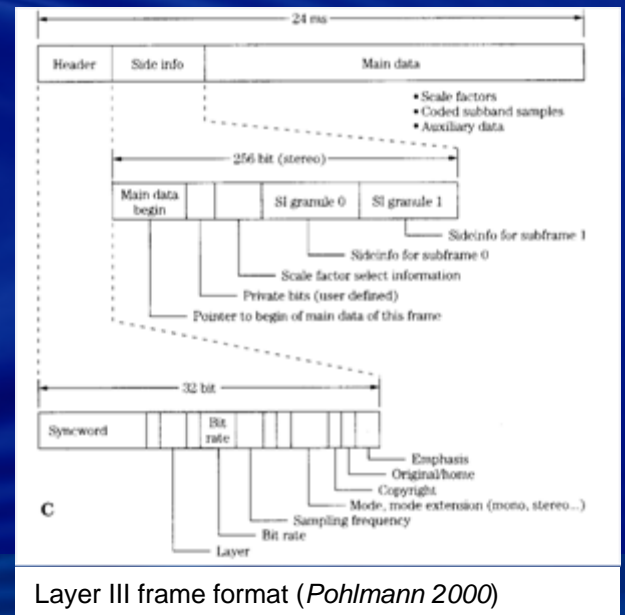
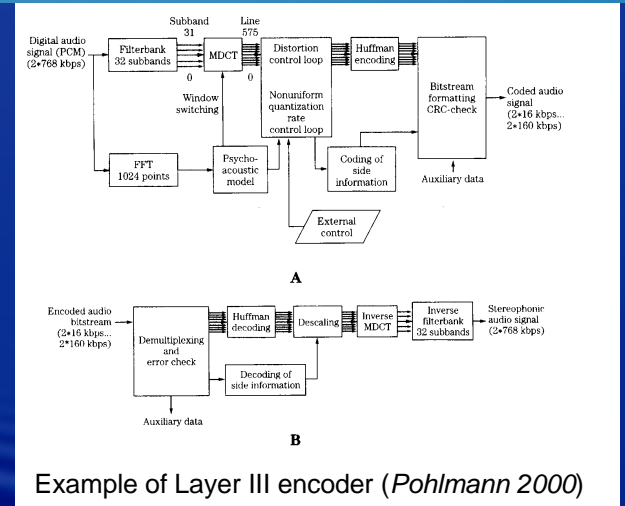
Header		CRC	Bit allocation	SCFSI	Scalefactors	Subband	Samples	AD		
A) 12 bit sync	16 bit	16 bit	Low subbands	00	6 bit linear	Gr0	Gr11	Auxiliary data field length not specified		
			4 bit linear						01	12 granules [Gr] of 3 subband samples each.
			Mid subbands						10	
B) 20 bit system info	16 bit	16 bit	3 bit linear	11	2 bit	3 subband samples correspond to 96 PCM audio input samples.	Auxiliary data field length not specified			
High subbands			2 bit linear							

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MPEG-1 Layer III (MP3)

- From 32 to 320 kbps
- Improvements:
 - Finer psychoacoustics model
 - Alias reduction (MDCT filters)
 - Nonuniform quantization
 - Entropy coding
 - Adaptive block size
- Only Layer with patents
- Inspired by:
 - ASPEC (audio spectral perceptual entropy coding)
 - OCF (optimal coding in the freq. domain)



Joint Stereo Coding

■ Intensity coding

- Sum of left/right channels
- Coding of the sum and of left/right scale factors
- Usually only for high-frequency subbands
- Efficient for redundant audio channels

■ MS (mid/side) stereo coding


- Sum and difference of left/right channels
- Coding of the two values

■ Stereo masking

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Audio Quality

Comparison with CD-quality at 48 kHz (16 bit - 1.412 Mbps) 



■ Layer I:

- No perceptual difference for 384 kbps (stereo) – 2:1 compression

■ Layer II:

- No perceptual difference for 256 kbps (stereo) – 4:1 compression

■ Layer III

- Increase of mean opinion score compared to Layer II at 256 kbps (stereo) for 128 kbps (stereo) – 8:1 compression  

Comparison between Layers

- Layers I and II are very similar
- Each Layer has its defined decoder design
- Encoding/Decoding complexity: Layer I/II
→ broadcasting
- Encoding/Decoding quality: Layer III
→ audio storage

Conclusion

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