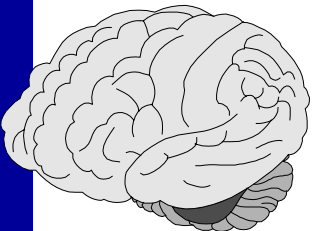
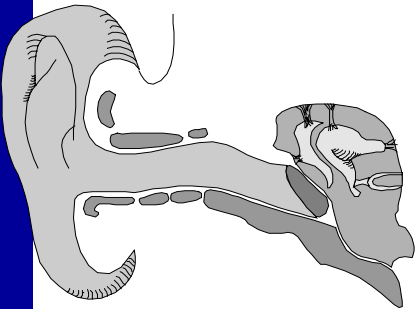
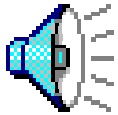


# Auditory principles in speech processing – do computers need silicon ears ?



\* with contributions by V. Hohmann, M. Kleinschmidt, T. Brand, J. Nix, R. Beutelmann, and more members of our medical physics group

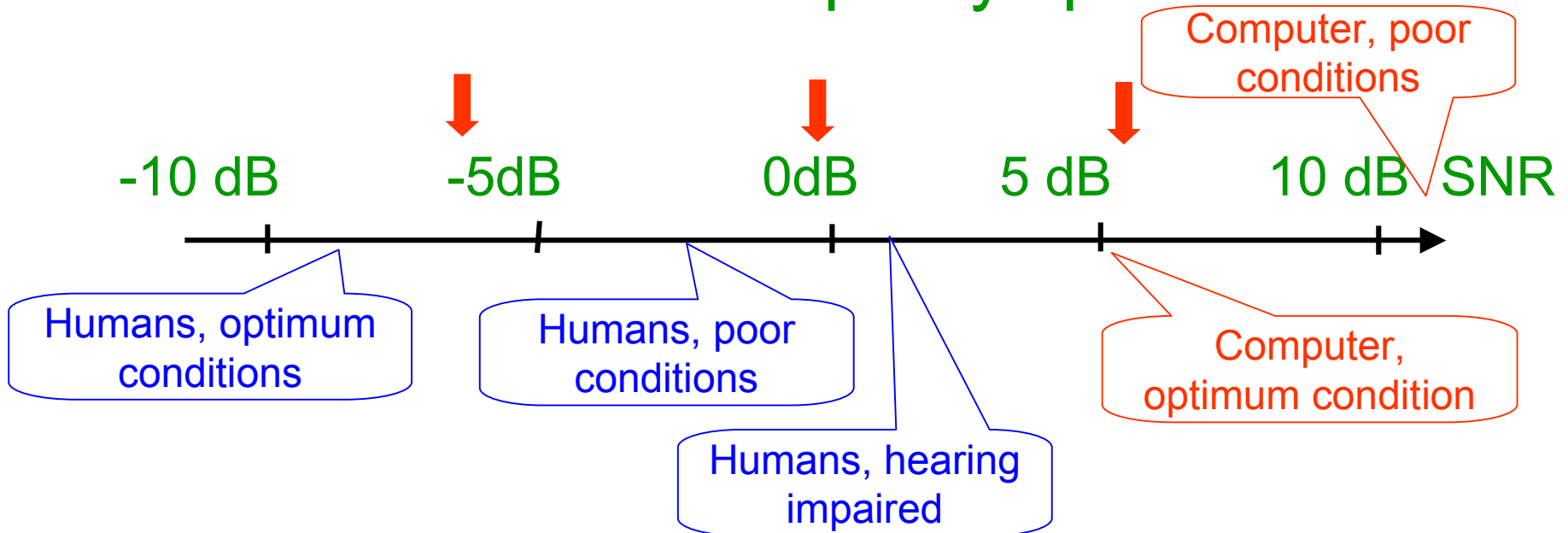
# A quick test of your Cocktail-party-processor....



Can you understand this sentence?

(Three repetitions, increasing signal-to-noise ratio)

Rachel has seven pretty spoons

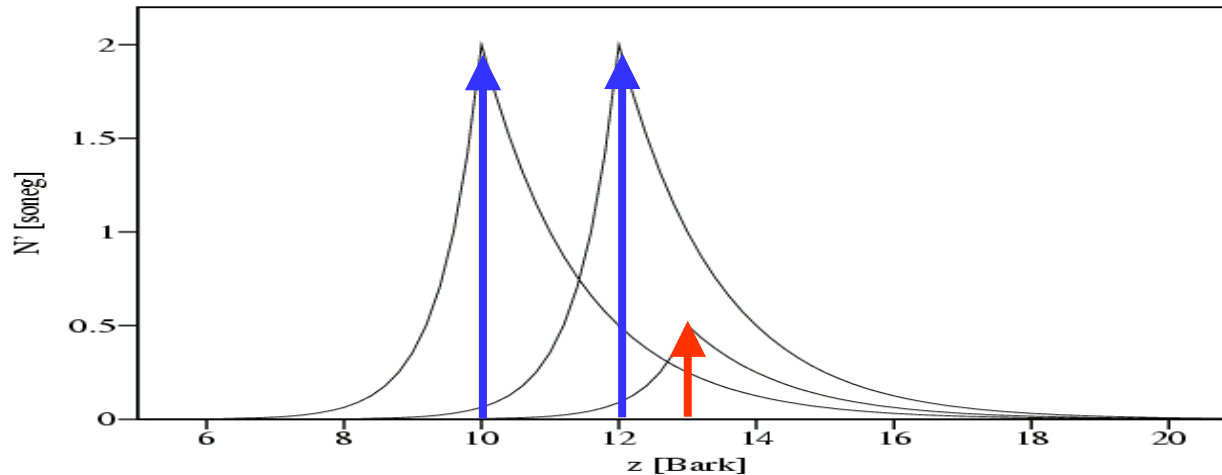
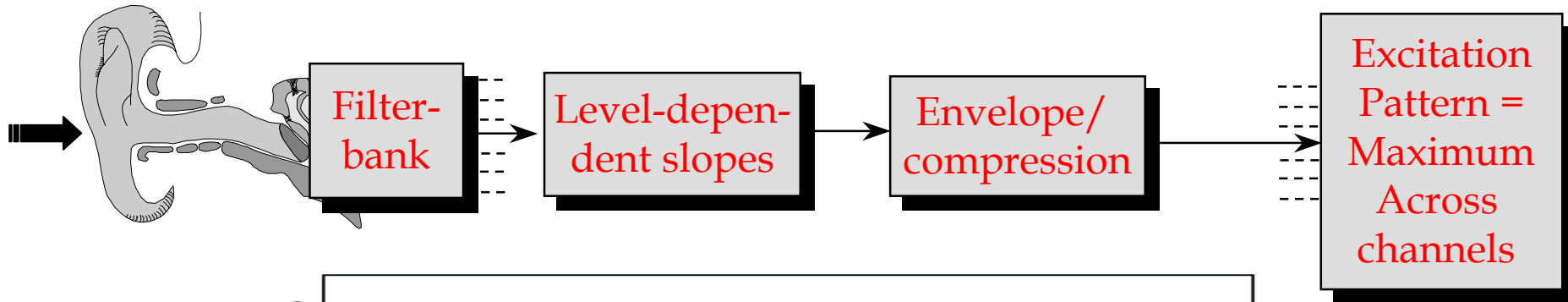


- Auditory principles already „in silico“
- Additional properties not yet exploited
- Auditory models
- Modulation processing
- Binaural information processing
- ...why it matters not only for hearing aids

# Auditory properties used in speech processing systems

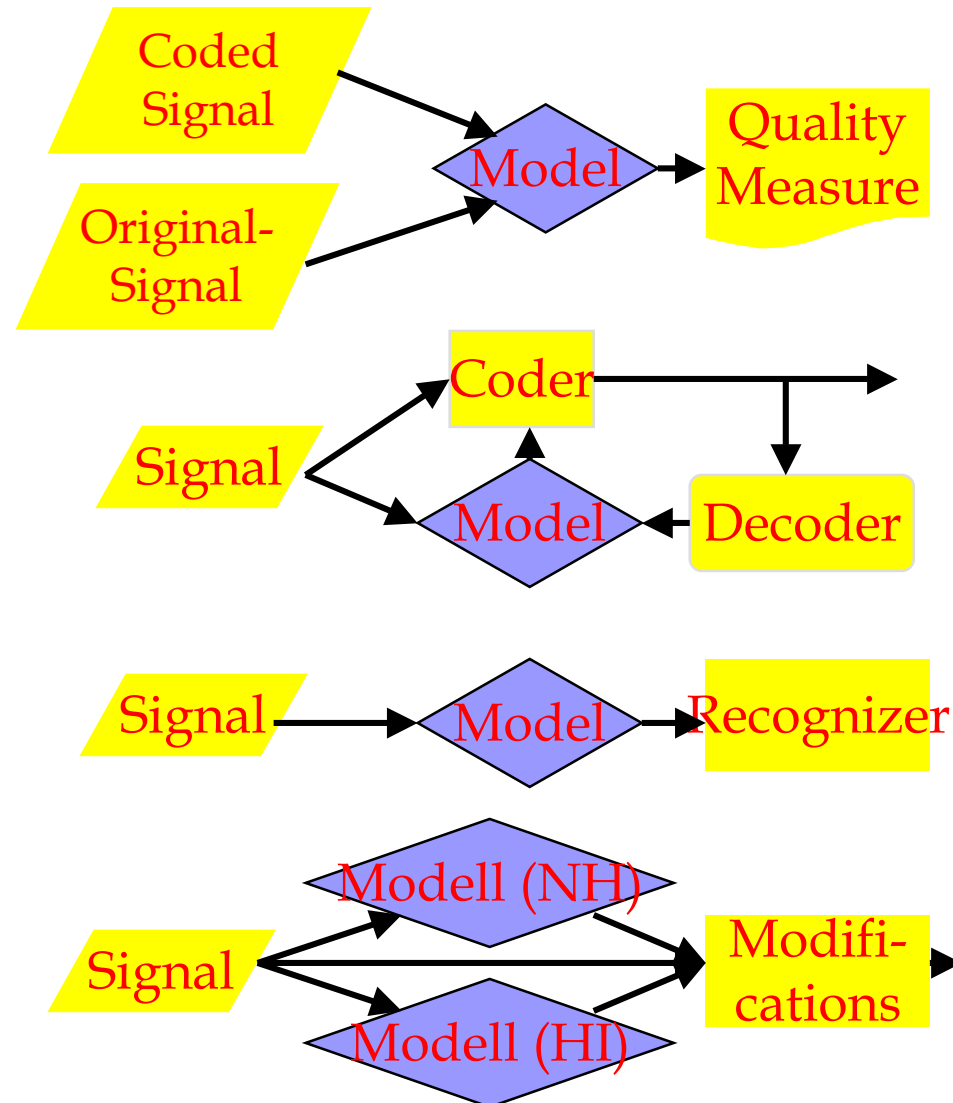
- Logarithmic/compressive intensity units (dB, loudness, cepstrum)
- Quasi-logarithmic Mel/bark-scale as frequency scale
- Linear predictive coding with quantization noise masked by speech (Schroeder & Atal)
- Masking model for HiFi-coding (MPEG, Brandenburg)
- Speech coding & Audio quality objective assessment (Beerends & Stemerdink)
- RASTA techniques for ASR (Hermansky & Morgan)

# Spectral Masking models



- Classical approach: Loudness model (Fletcher, Stevens, Zwicker & Fastl, Moore)
- Forward & backward masking produce temporal sluggishness

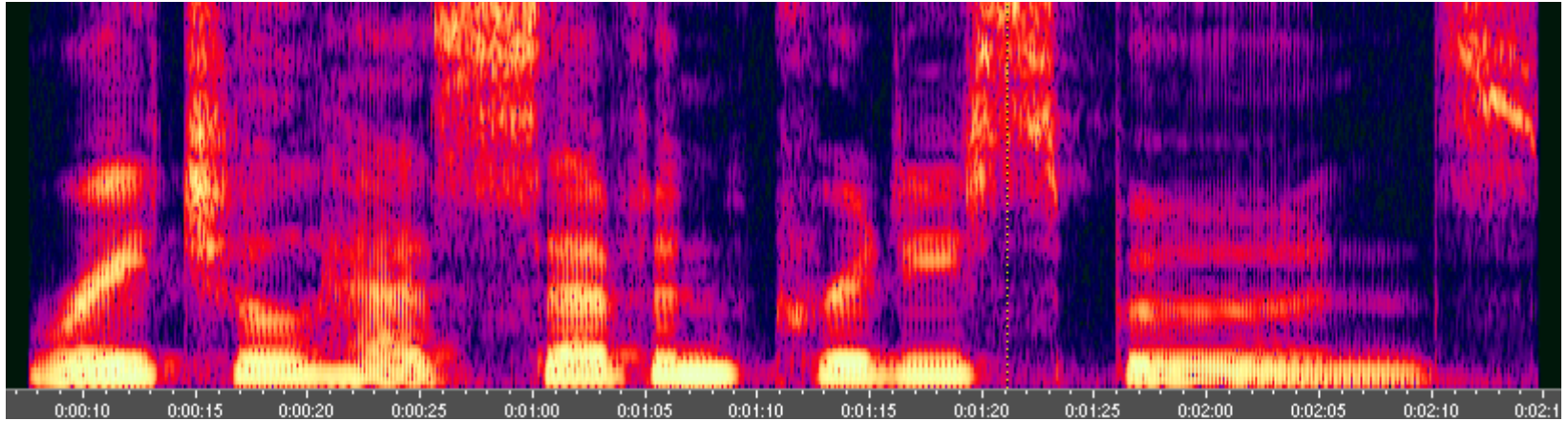
- Assessment of signal quality (Cellular phone networks,....)
- Signal coding (MP3, MiniDisc,..)
- Speech & pattern recognition
- Hearing aids



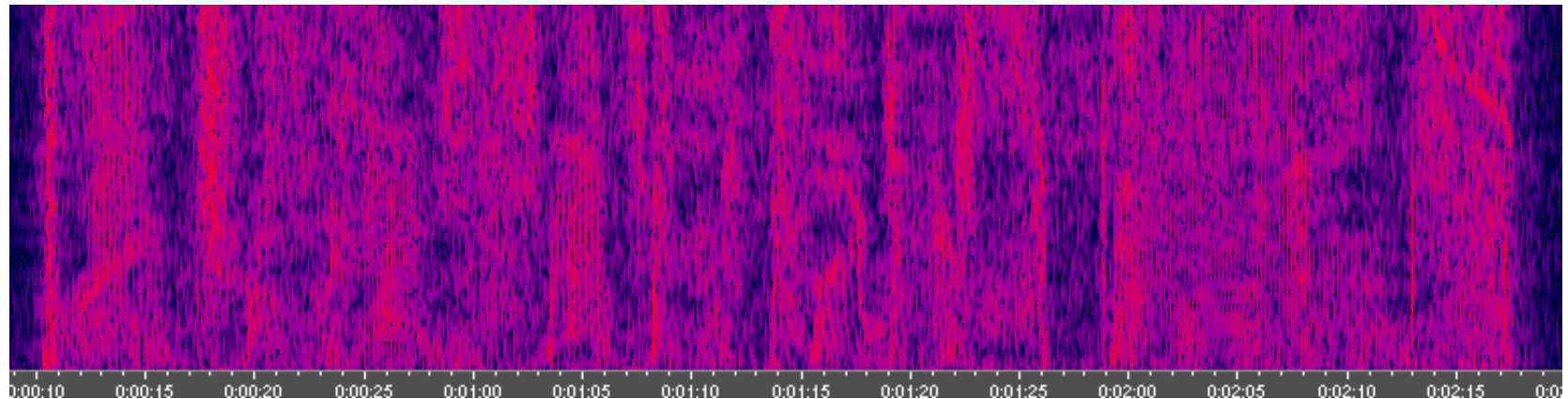
# What do we miss when applying classical models?

- ➔ Temporal processing
  - Modulation processing
  - Spectro-temporal modulation processing
  - Binaural/spatial processing
  - Cognitive effects (interpolation, suppression)

# Speech perception without a spectrum?



Flat spectrum (phase only) speech using the Oldenburg sentence test

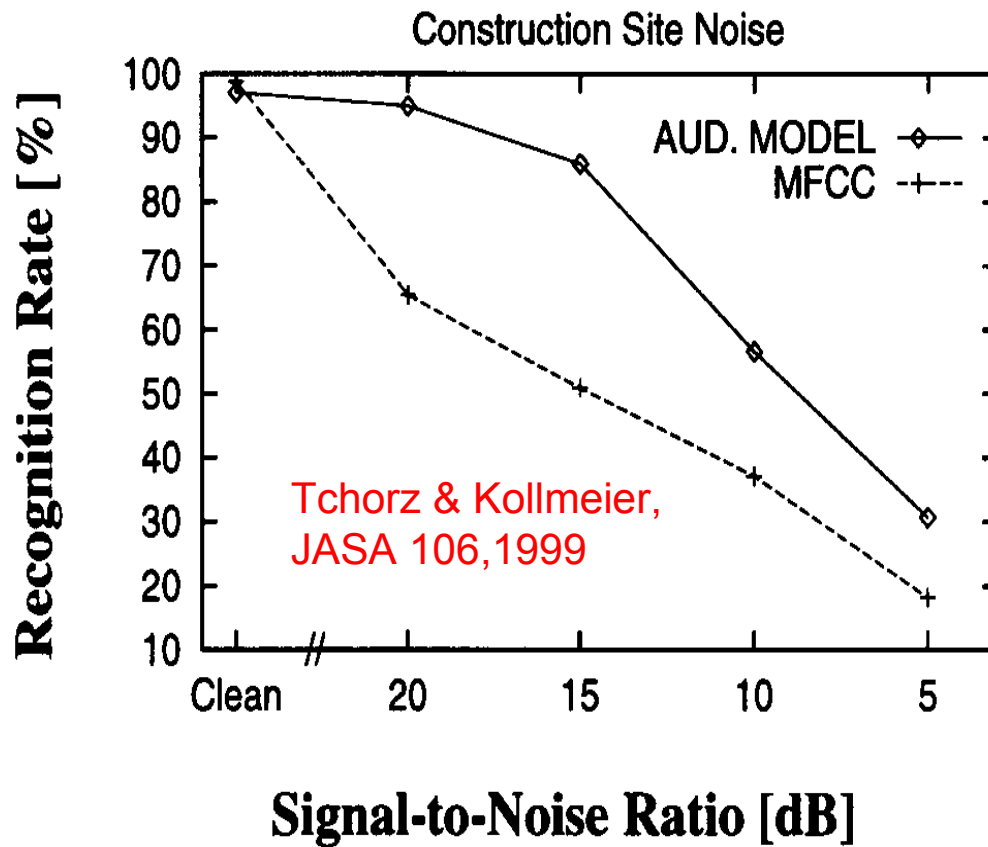
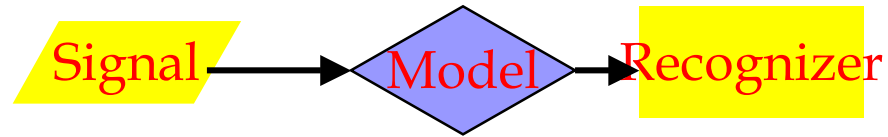


Ear performs temporal analysis



# Temporal processing used for robust speech recognition

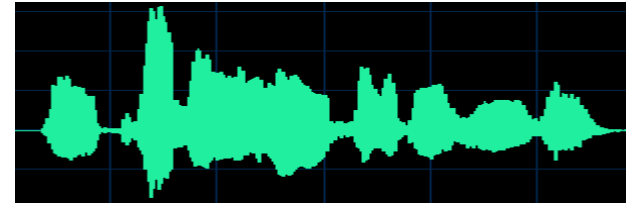
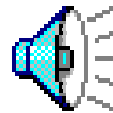
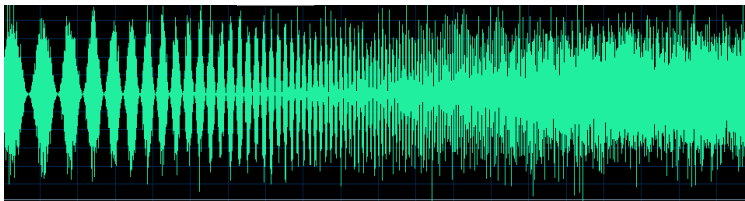
- Hermansky & Sharma: TRAPS
- Perception model



„Auditory front end  
for speech  
recognizer“:  
Robustness  
against noise

# What do we miss when applying classical models?

- Temporal processing
- ➔ **Modulation processing**
- Spectro-temporal modulation processing
- Binaural/spatial processing
- Cognitive effects (interpolation, suppression)



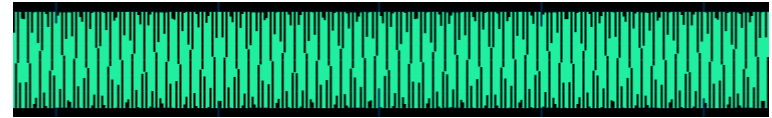
Sinusoidally Amplitude modulated noise

Spoken sentence

# Asymmetry in masking: Which is the „better“ masker?

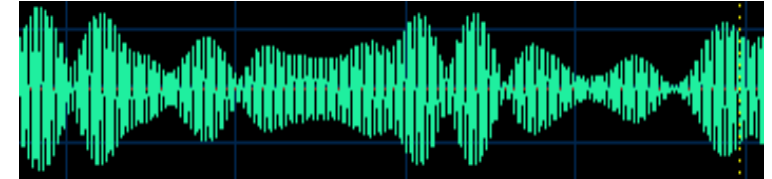


Tone 2 kHz



Narrow-band noise

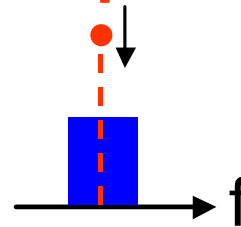
2 kHz, 256 Hz bandwidth



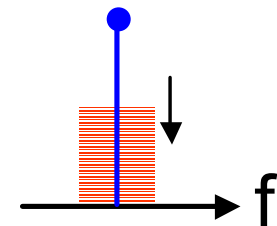
## Count the audible steps!



Tone in steps masked  
by continuous noise



Noise in steps masked  
by continuous tone



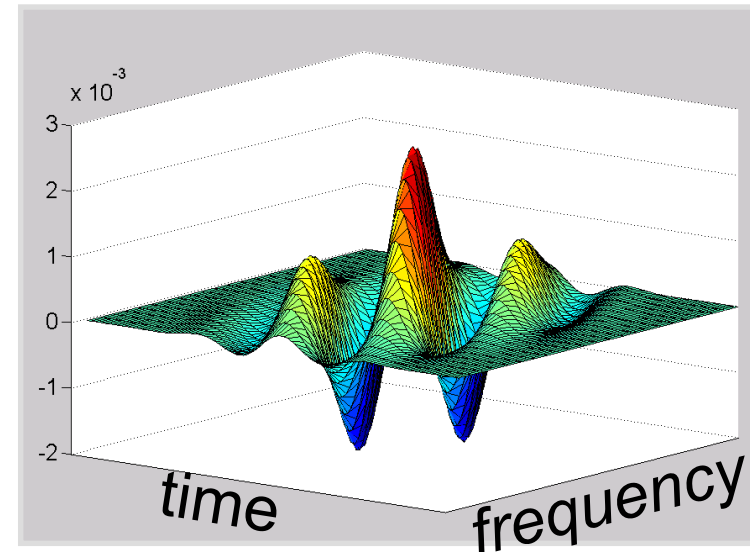
➔ Ear performs detailed envelope analysis (modulation spectrum)

# What do we miss when applying classical models?

- Temporal processing
- Modulation processing
- **Spectro-temporal modulation processing**
- Binaural/spatial processing

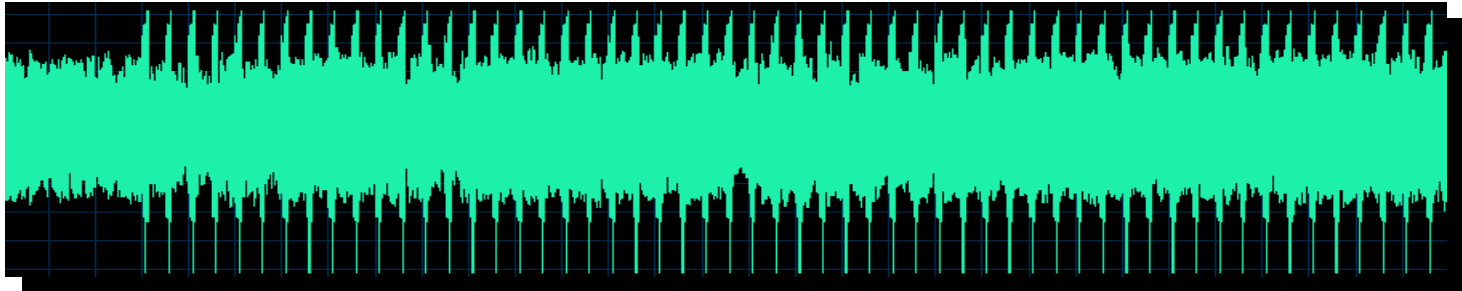
Gabor spectro-temporal  
feature

➔ **Cognitive effects**

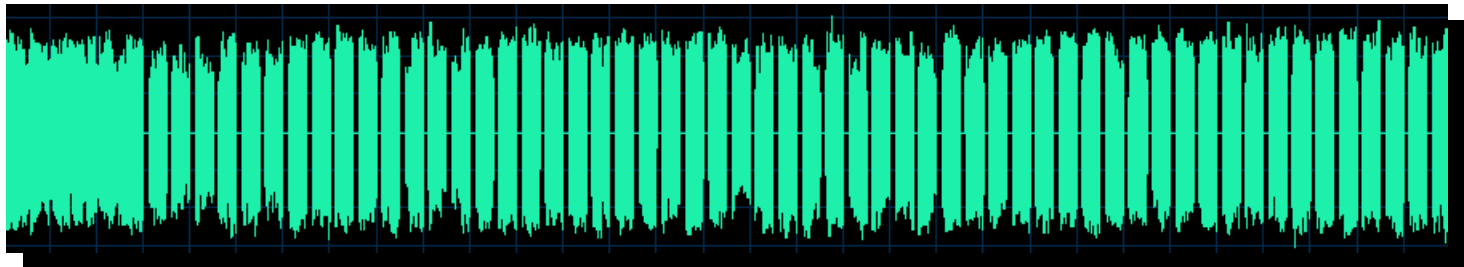


# Continuity illusion: Can we trust our ears?

Music + pauses + noise (+6dB)



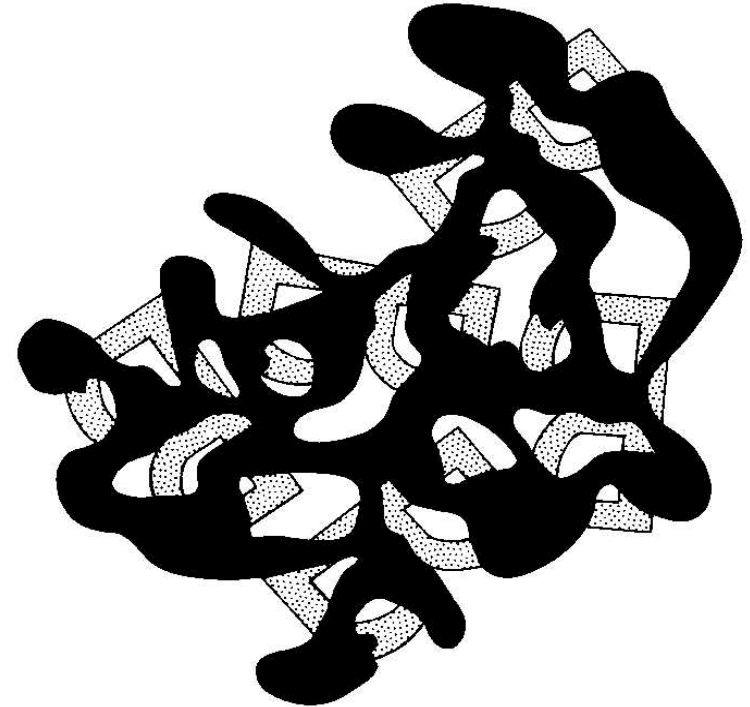
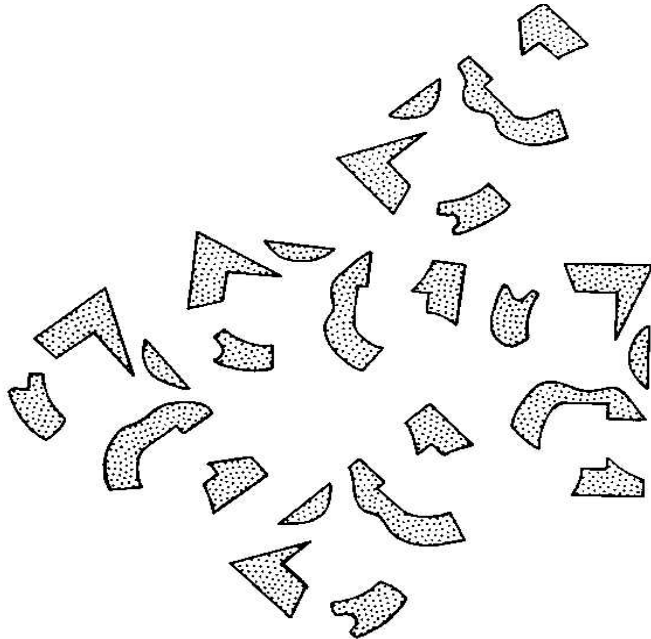
Do you hear ongoing music?



Music & pauses (500/125 ms)

➔ top-down processing by our brain

# Visual analogy

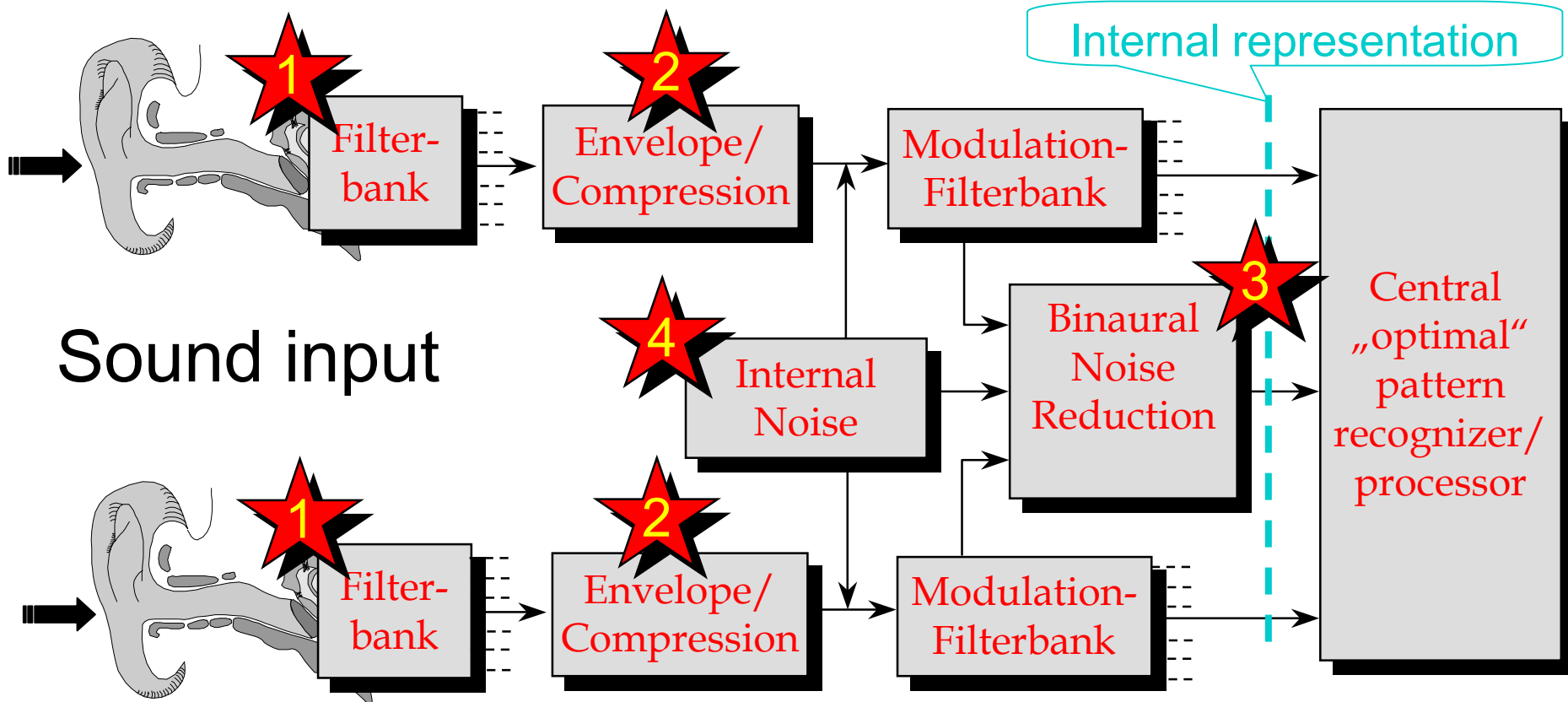


Bregman's Bs

# What do we miss when applying classical models?

- Temporal processing
- Modulation processing
- Spectro-temporal modulation processing
- Binaural/spatial processing
- Cognitive effects (interpolation, suppression)

How can we quantify these effects and put them to work in speech processing?



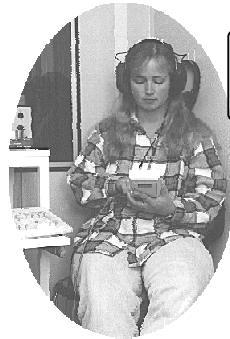
Model of the „effective“ processing in  
the auditory system & *Impairments*



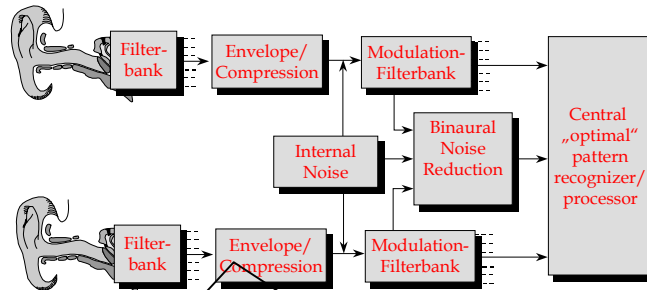
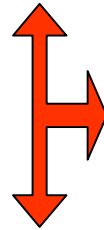


# Approach: Analysis by model & simulation system

Acoustical scene

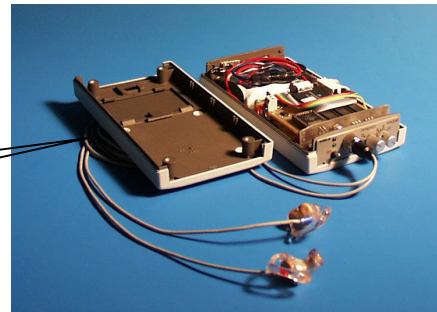


Human listener



Auditory model

Hearing Aid/  
computer



Test result



Predicted Test result



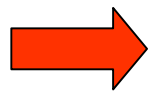
System performance

Auditory System exploits all available acoustical cues, employing

- Lossy Front End (quantified by auditory model with information compression):  
acoustical input → „internal representation“
- Perfect Back End: central pattern recognition (limited only by „internal noise“)

Technical „copy“ of auditory front end  
→ yields near optimum performance of technical system

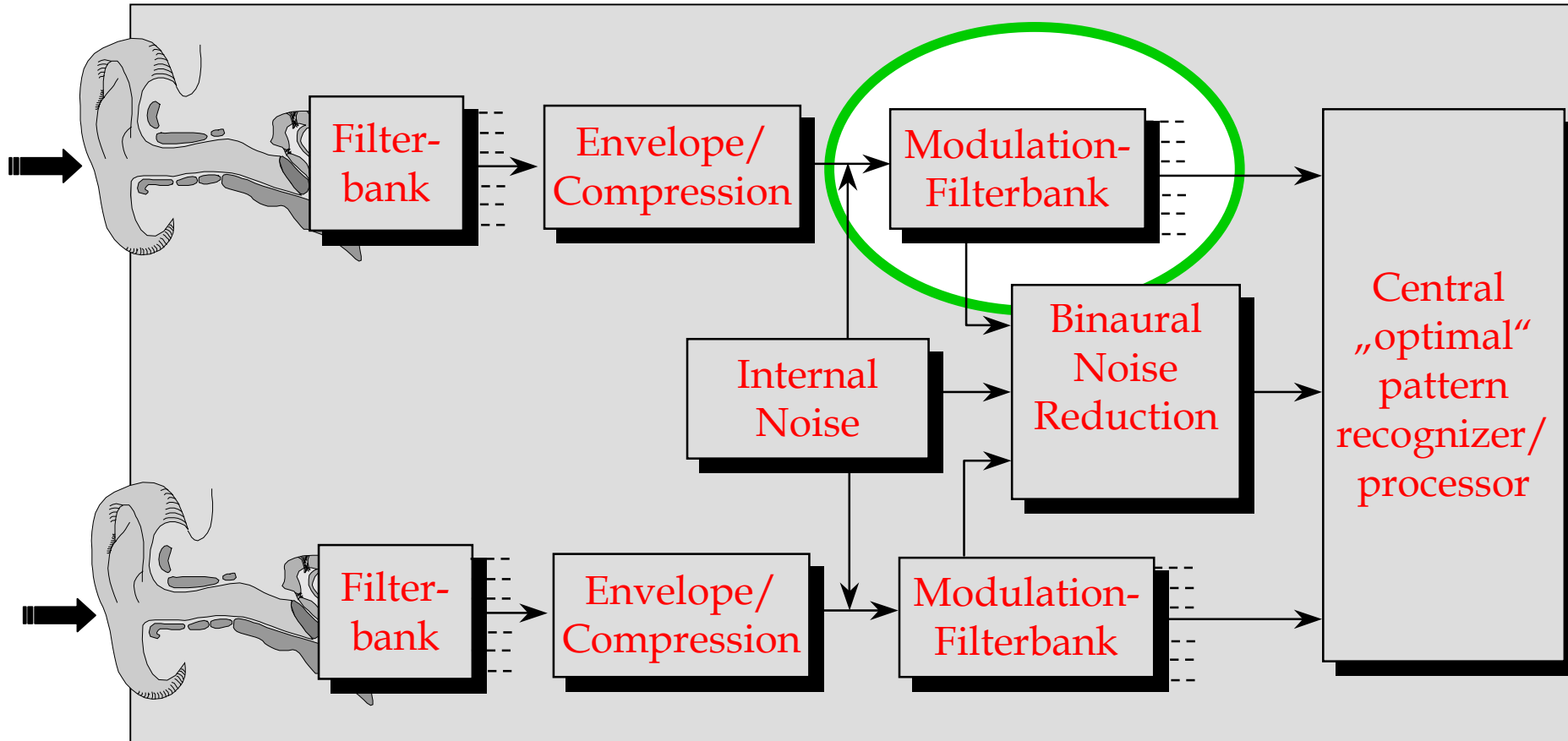
- Auditory principles already „in silico“
- Additional properties not yet exploited
- Auditory models



## Modulation processing

- Binaural information processing
- ...why it matters not only for hearing aids

# Model framework: Modulation frequency analysis

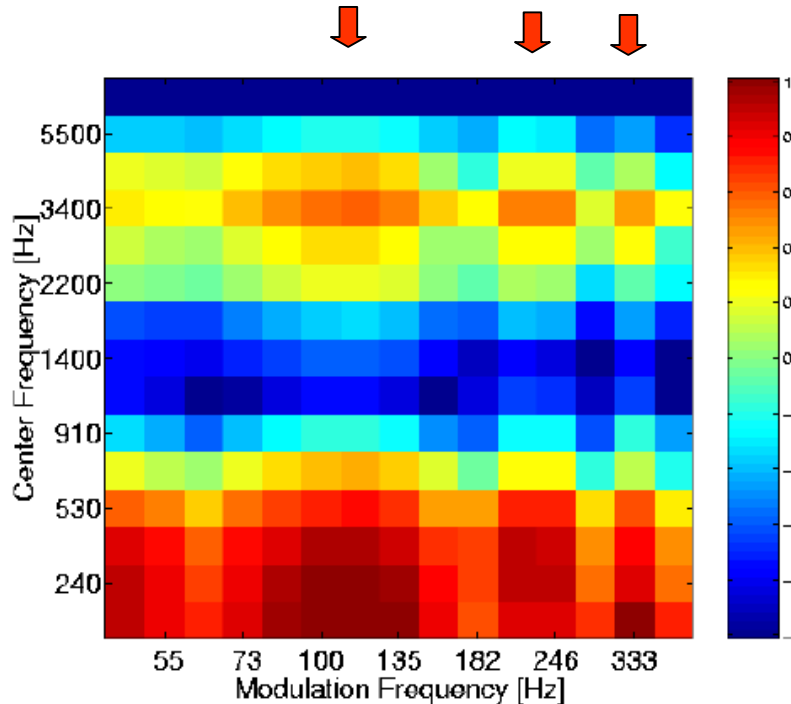


Model of the „effective“ processing in  
the auditory system

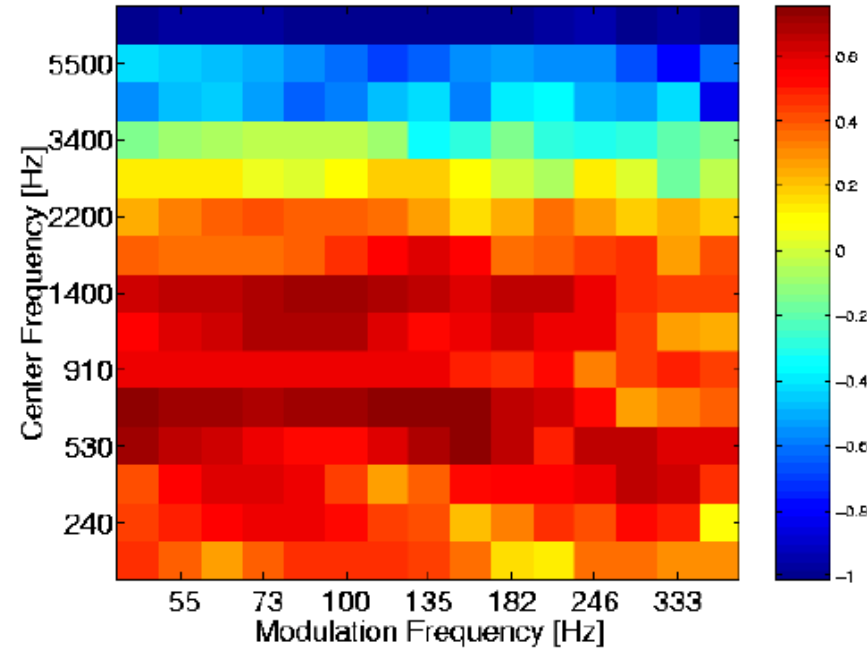
# Modulation frequency selectivity

- **Modulation-map in the auditory system**  
(Langer & Schreiner)
- **Psychoacoustics: Modulationfilterbank**  
(Dissertation Dau, Dau et al., JASA 1997, Dissertations Verhey, Derleth, Ewert)
- **Signal processing, noise reduction** (Kollmeier & Koch, JASA 1994)
- **Advantage: Separation of different auditory objects covering the same frequency region**

(voiced) speech

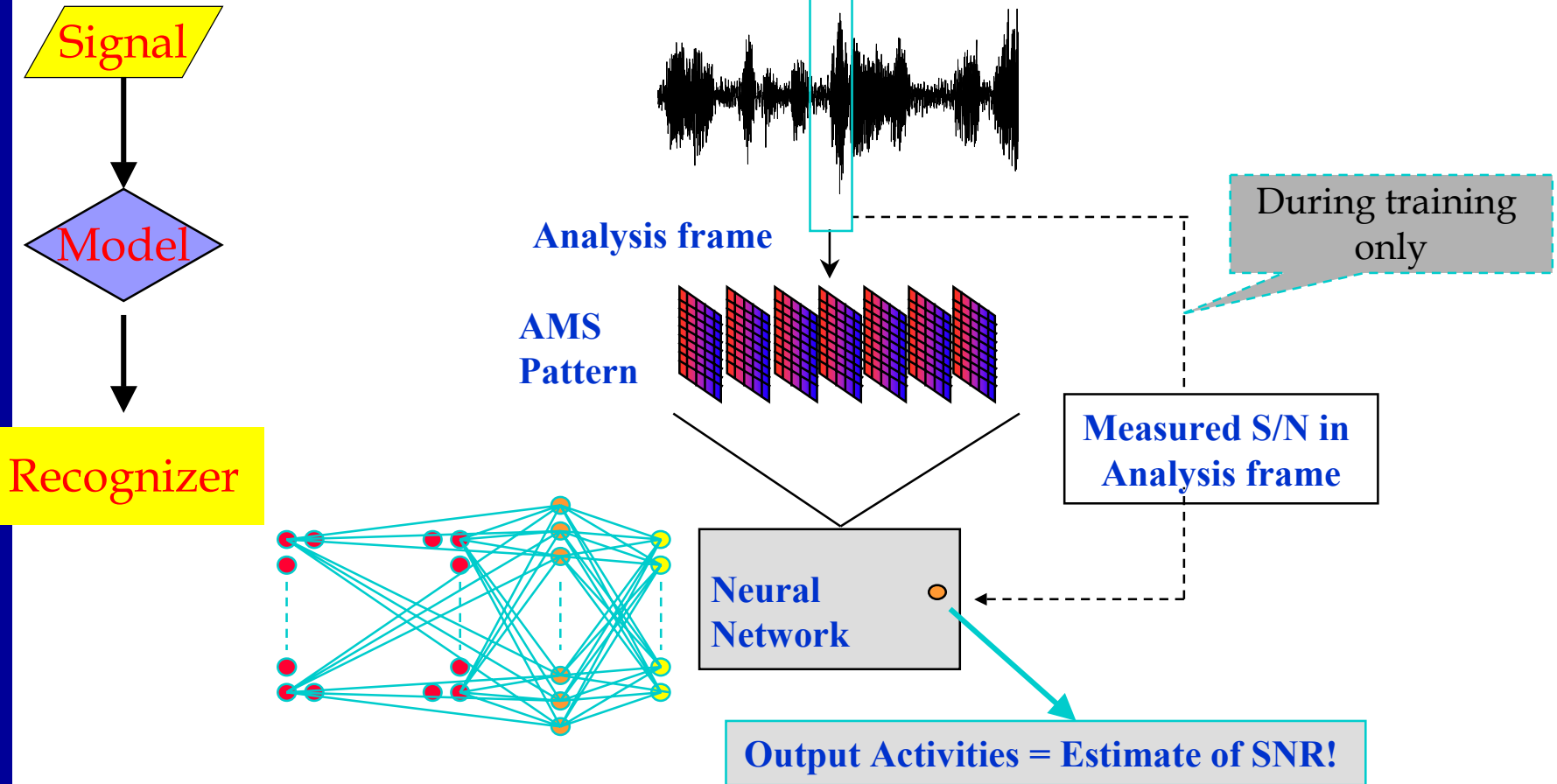


speech simulation noise



➔ Speech shows joint distribution in frequency/modulation frequency domain

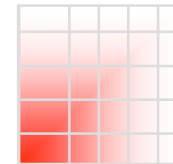
# SNR Estimation from Modulation spectrograms



Speech-to-Noise Ratio estimate either  
broadband or multiple narrowband

## ● Estimation error based on

- Full 2-dim distribution



5.2 dB

- modulation spectrum



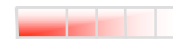
6.6 dB

- bark spectrum



7.6 dB

- combination of both



+



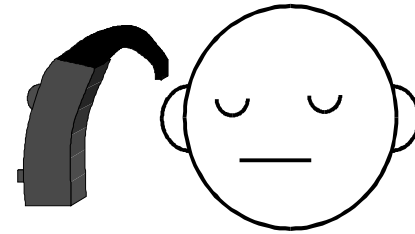
5.8 dB

➔ Joint distribution of modulations and spectrum required!



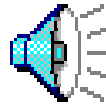
# Monaural Noise reduction using Amplitude Modulation cues

## Suppression of a fluctuating background noise using AMS



Industrial noise with speech

unprocessed



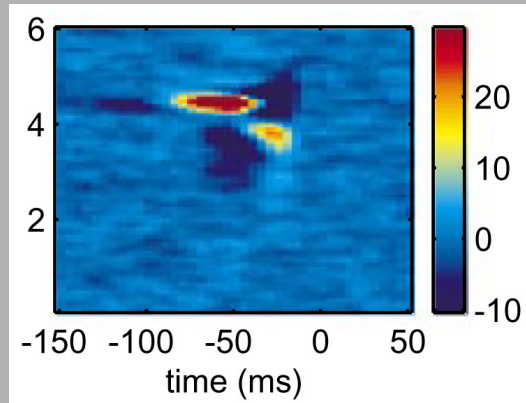
processed

Improves speech recognizer in noise (Tchorz & Kollmeier, 2002)

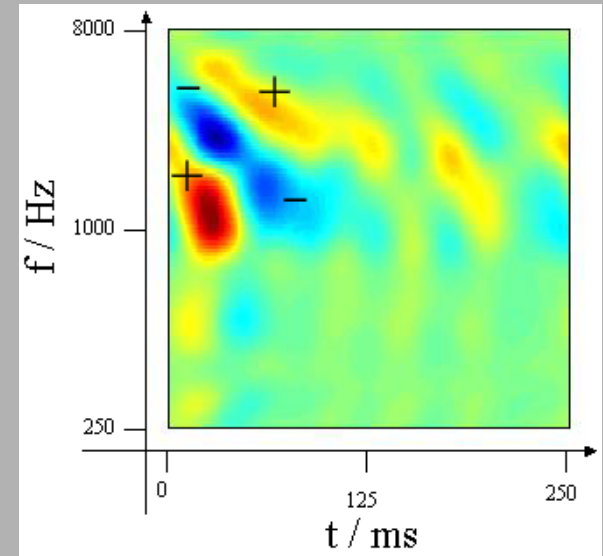


Modulation frequency analysis is important for speech perception & promising for speech processing

## Receptive fields of cortical neurons



DeCharms et al. (1998)

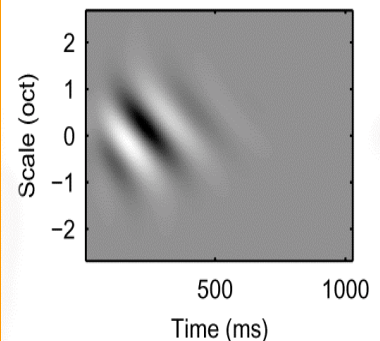


Depireux  
et al. (2000)

Indications for  
spectro-temporal  
feature extraction

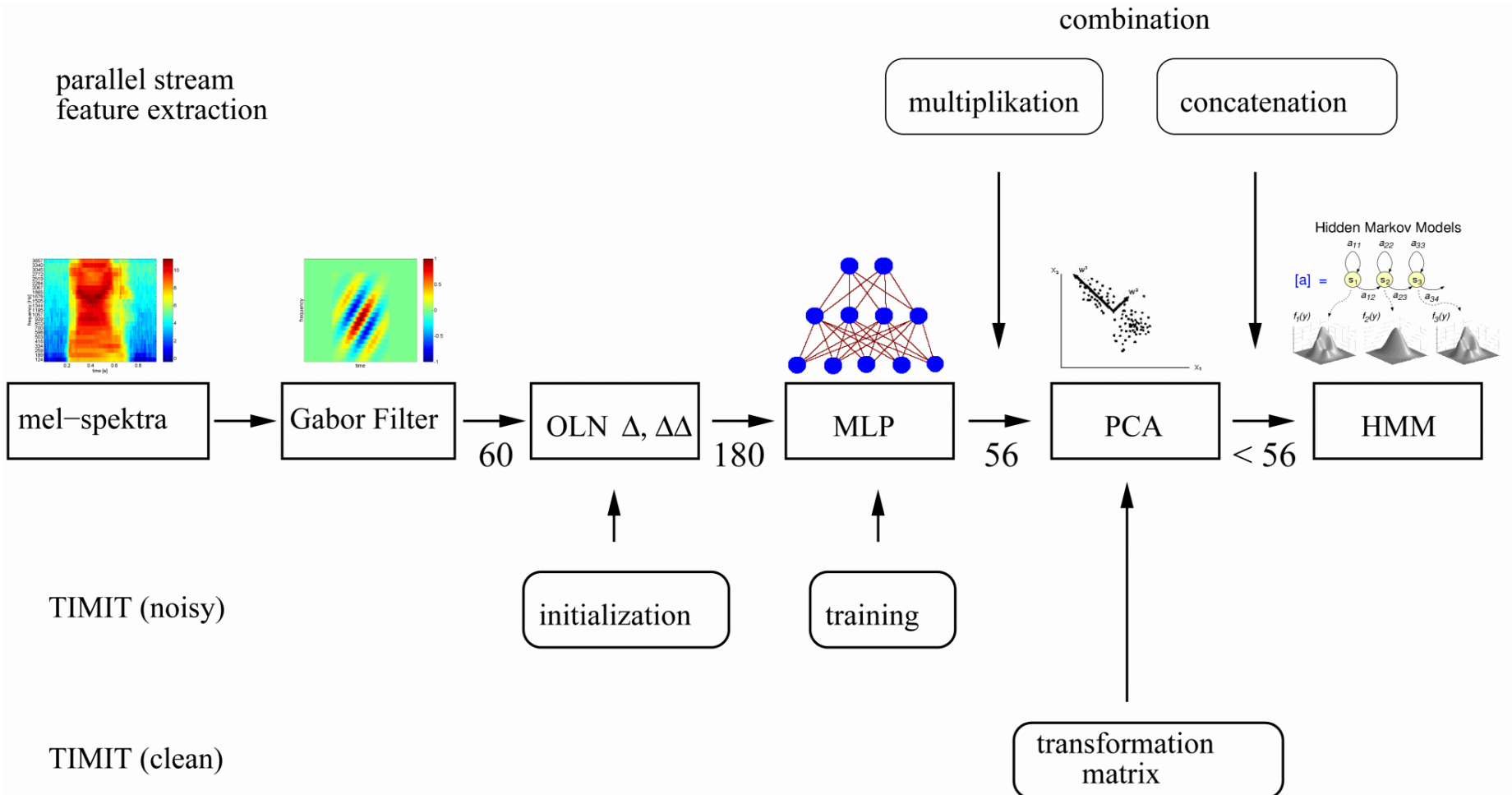
Model of  
modulation  
perception

Chi et al. (1999)

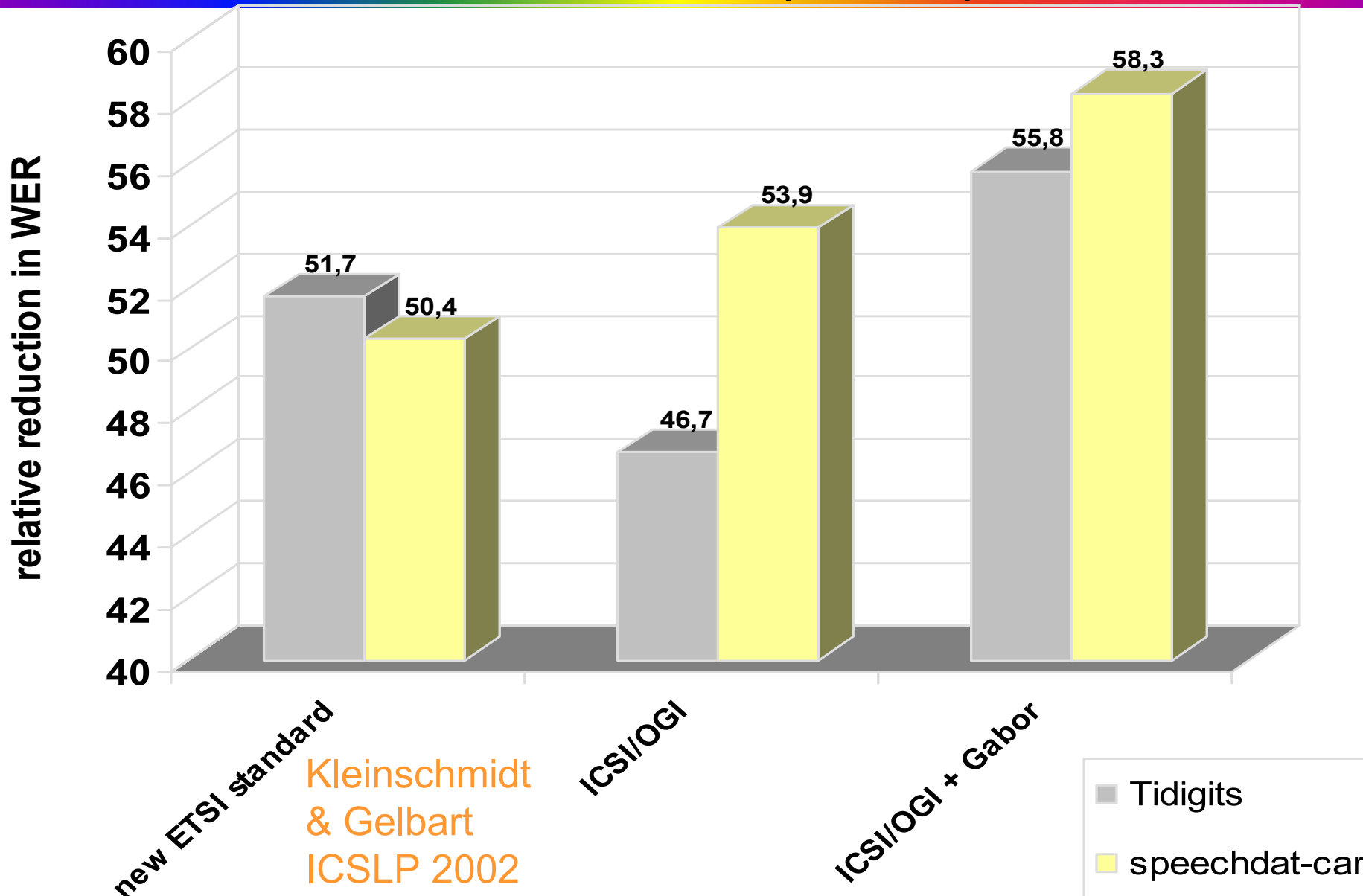


# Gabor Tandem recognizer

More details: M. Kleinschmidt: 'Localized Spectro-temporal Features for ASR'  
Eurospeech Session SThBb - Time is of the Essence, Thursday 10am, Room 2

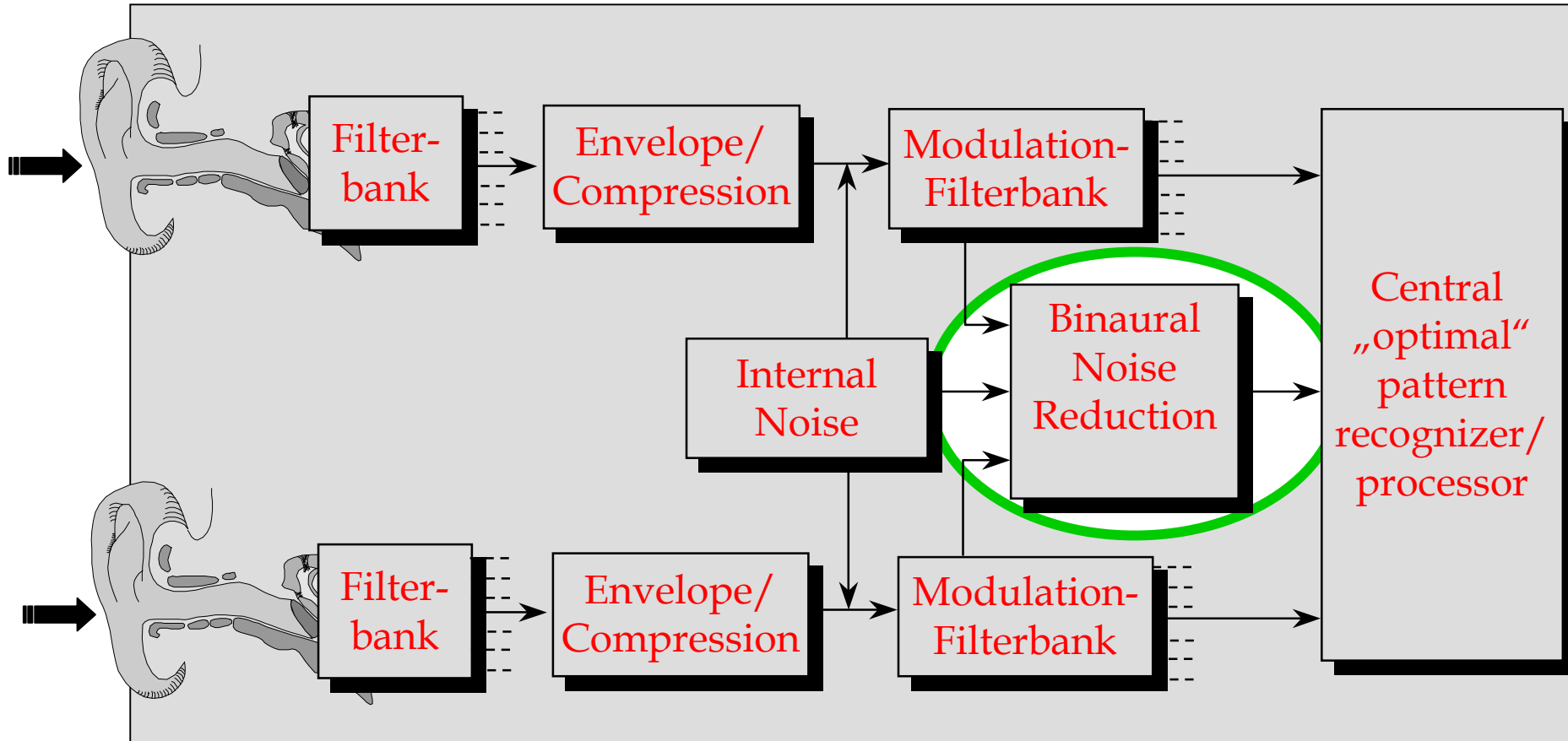


# Aurora: relative reduction in word error rate (WER)



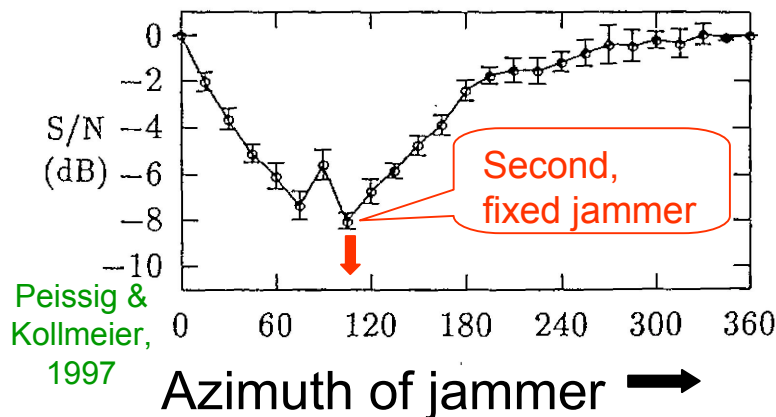
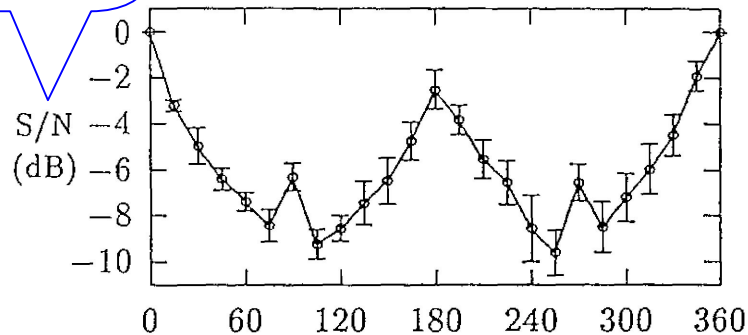
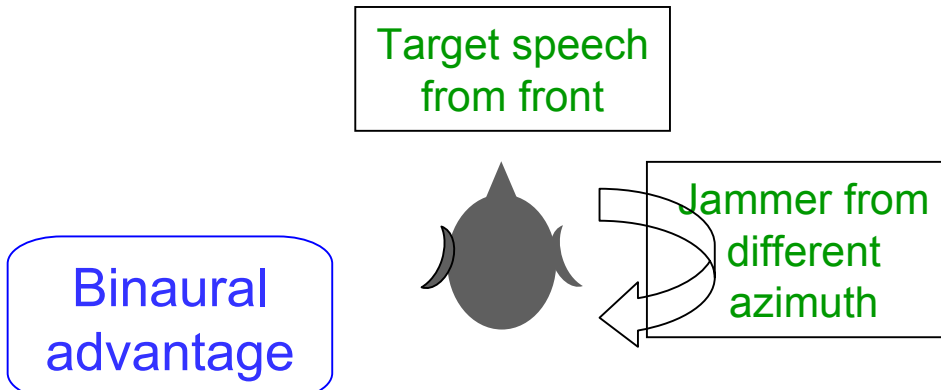
- Auditory principles already „in silico“
- Additional properties not yet exploited
- Auditory models
- Modulation processing
- ➔ **Binaural information processing**
  - ...why it matters not only for hearing aids

# Model framework: Binaural noise reduction



Model of the „effective“ processing in  
the auditory system

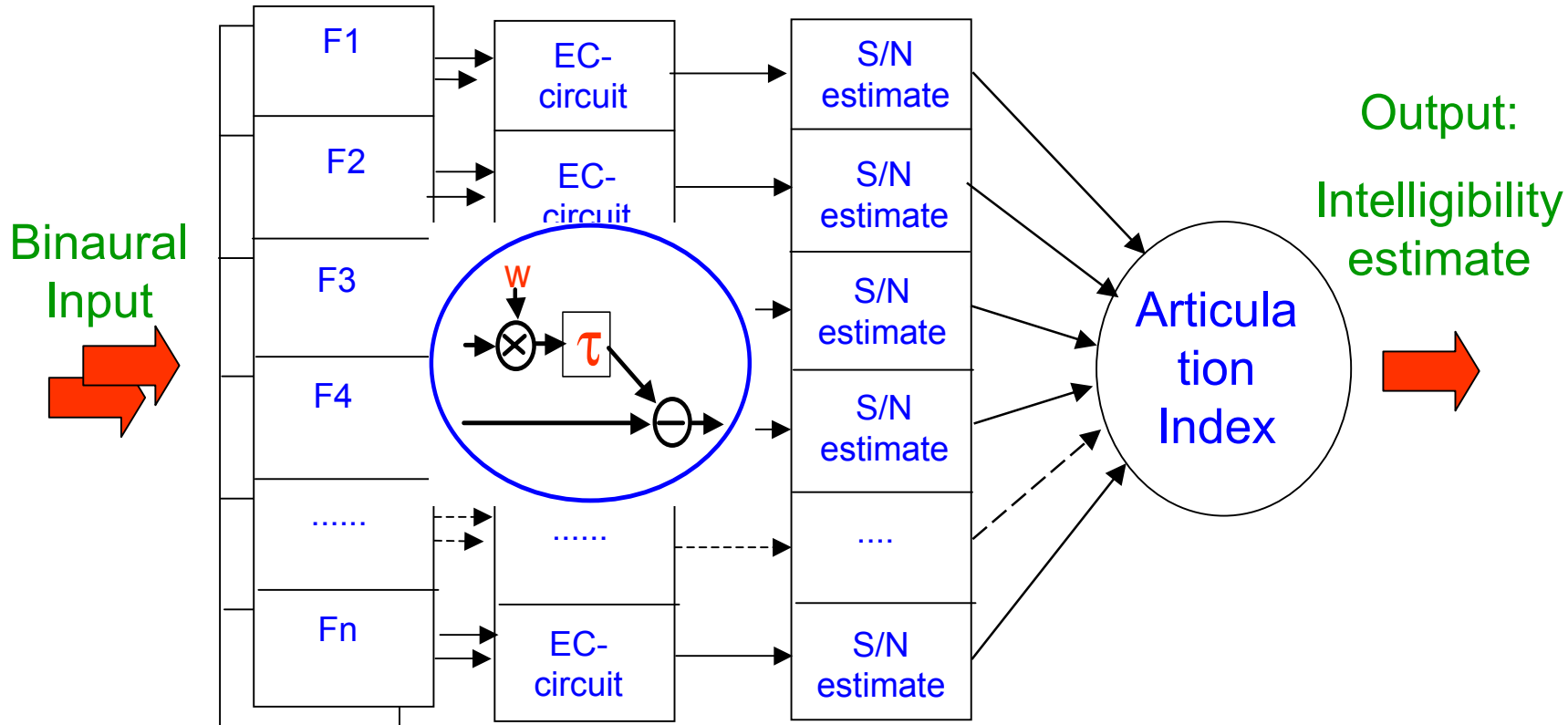
# Speech Reception Threshold for different spatial arrangements



- One continuous jammer provides maximum effect
- Second, opposite jammer can not be cancelled simultaneously

➔ Binaural hearing operates like 2-sensor adaptive beamformer

# v.Hövel-model ('84) model



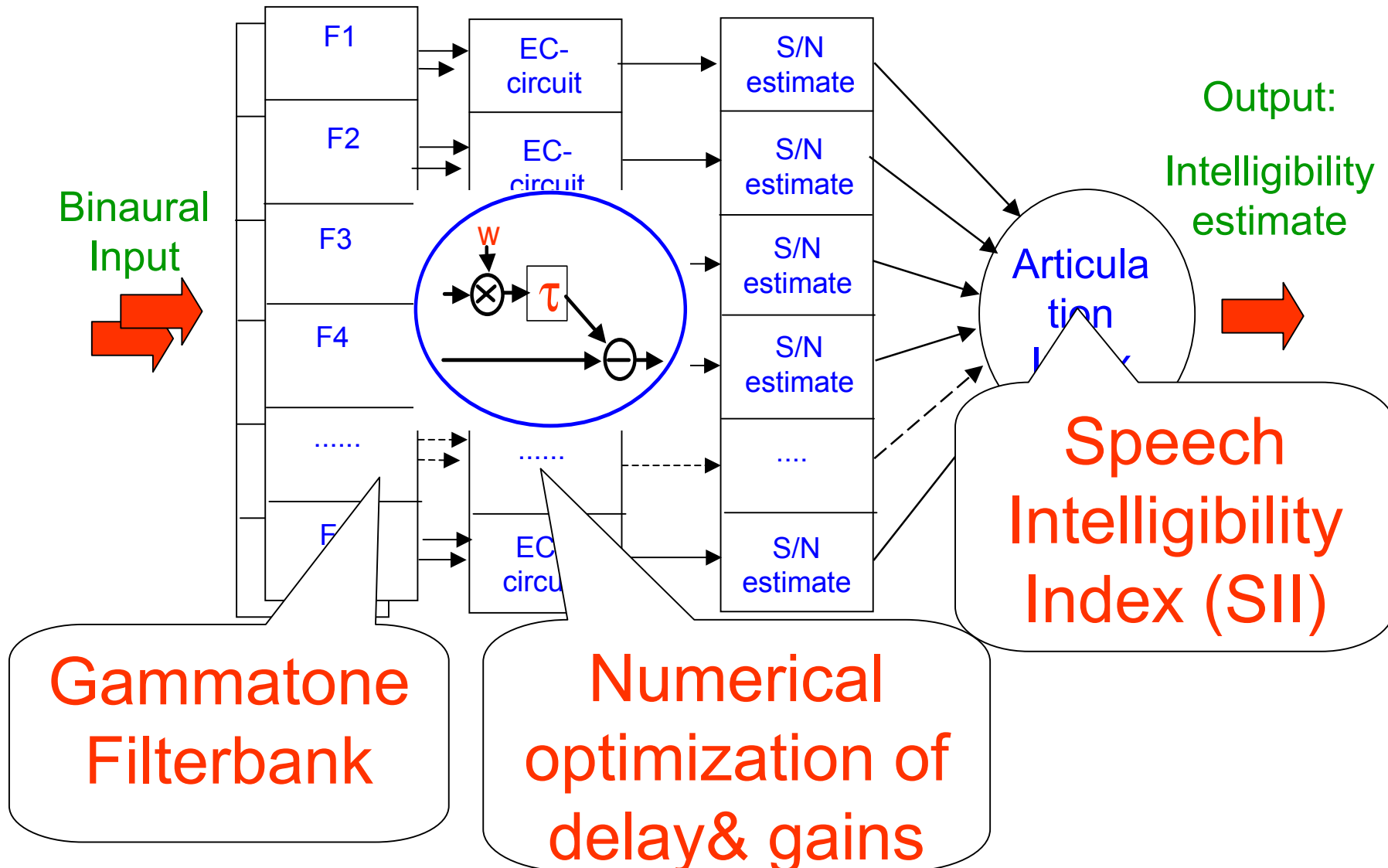
Filterbank

Noise  
reduction

Zur Bedeutung der Übertragungseigenschaften des Außenohres sowie des binauralen Hörsystems bei gestörter Sprachübertragung, Dissertation, RWTH Aachen

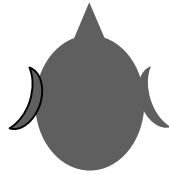


# v.Hövel-model ('84) model: modifications

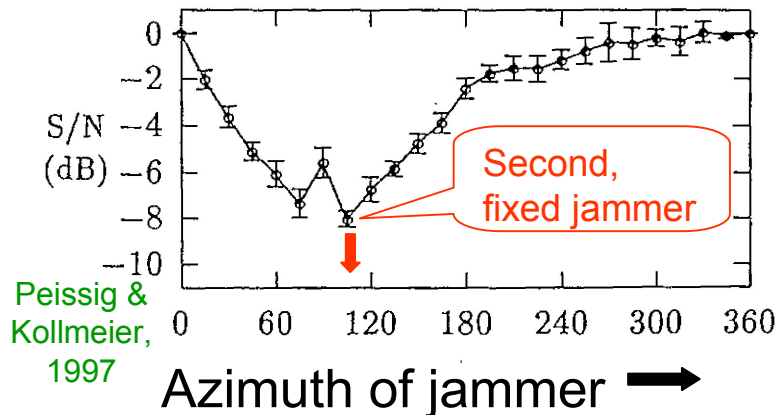
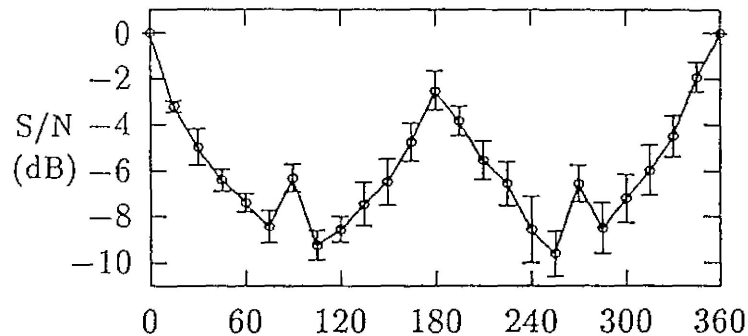
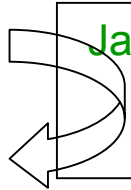


# Speech Reception Threshold for different spatial arrangements

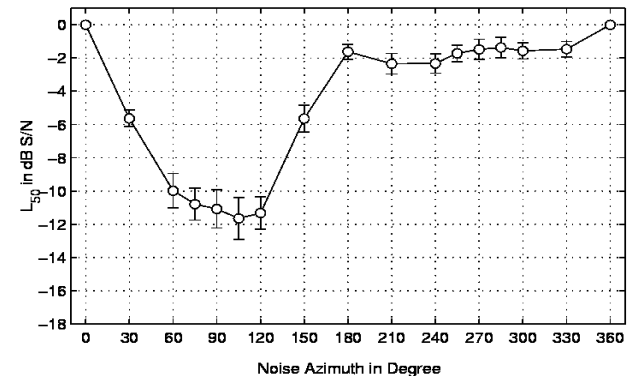
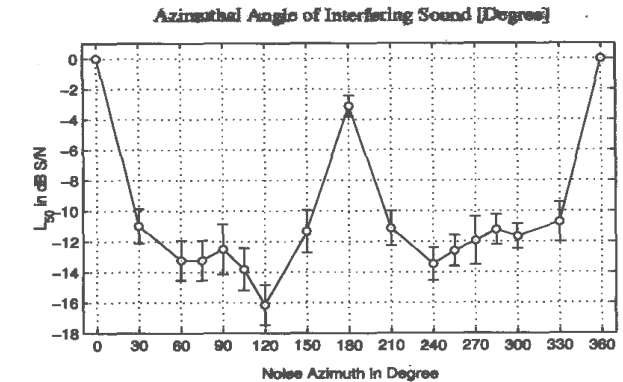
Target speech  
from front



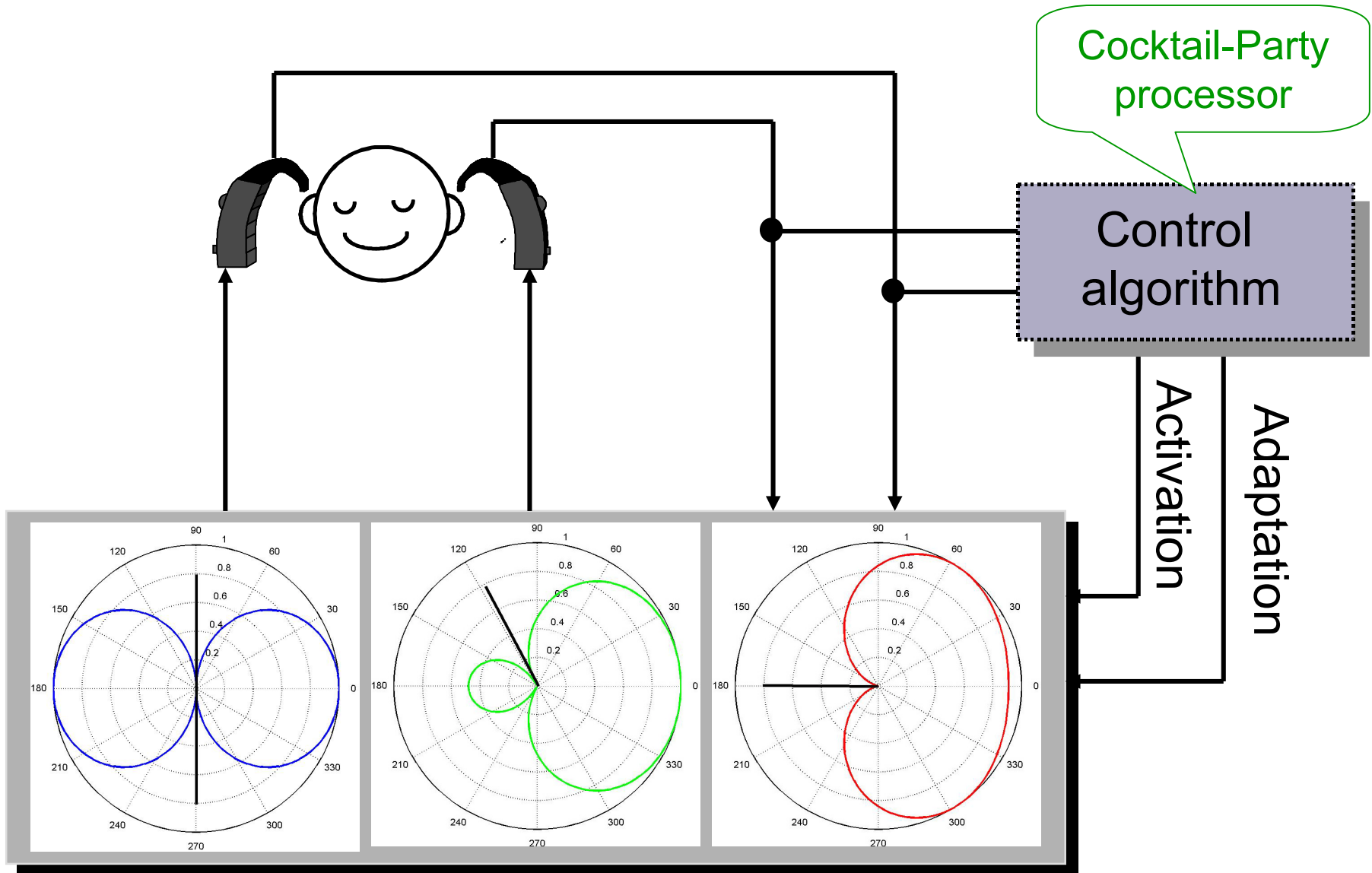
Jammer from  
different  
azimuth



Predictions by  
binaural model  
(R. Beutelmann, T.Brand)



# Corresponding binaural Beamformer hearing aid



# Performance of two-input „Cocktail party processors“

## Blind Sound Source Separation (Anemüller&Kollmeier, 2002)

- Mixture of two sources in a room
- Separation of first source

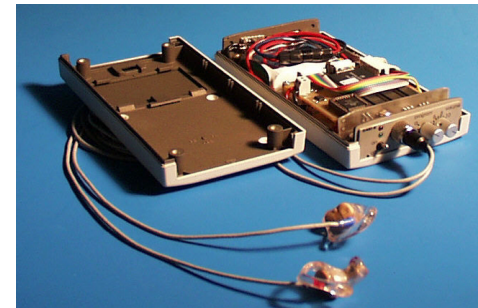
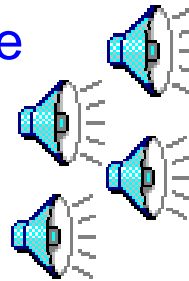


second source



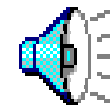
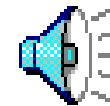
## Binaural situation-adaptive directional filter (Wittkop, 2000)

- One speaker in stationary noise
- One speaker from the front
  - + 3 interfering speakers
  - + Algorithmus



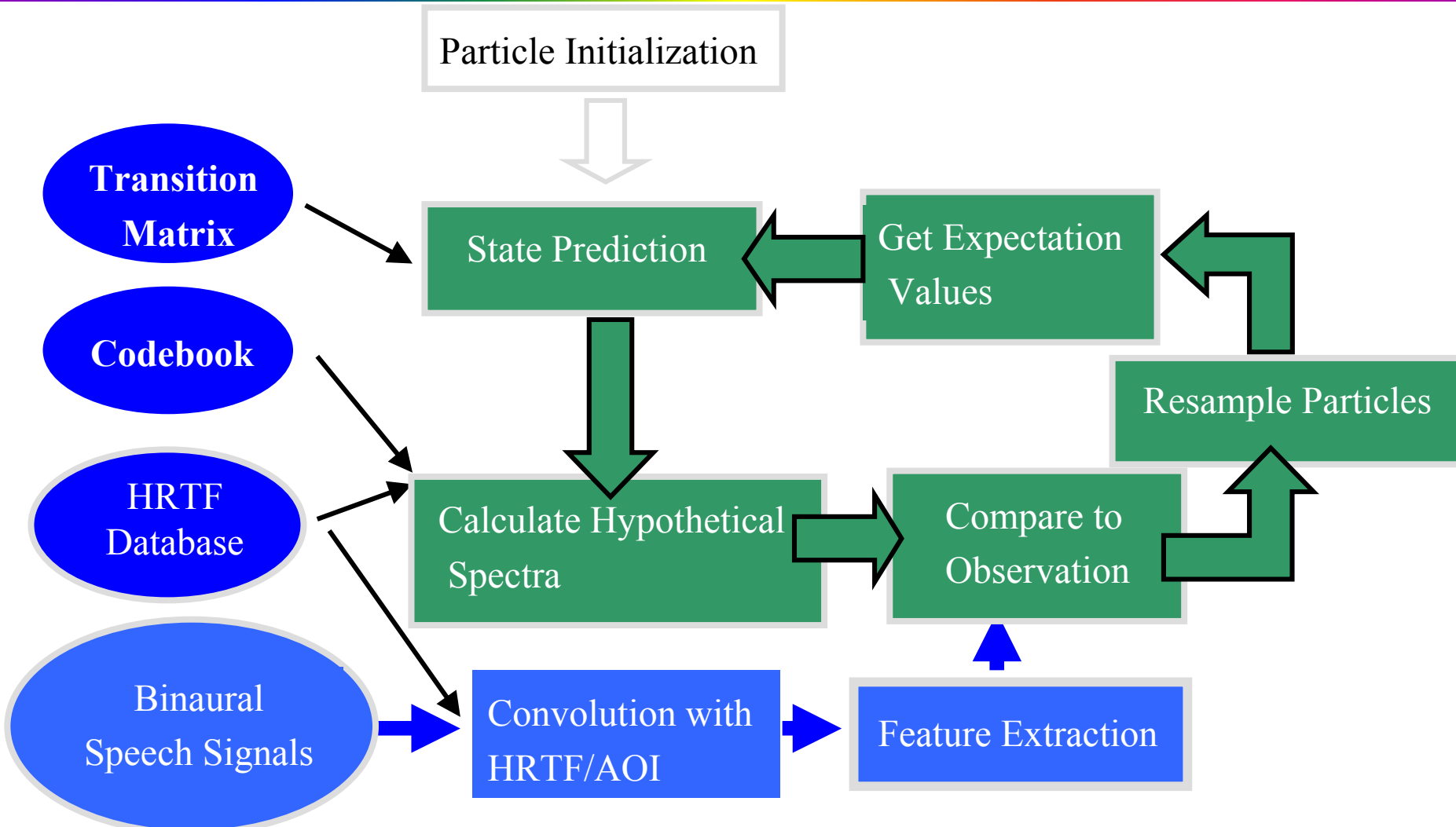
## Localization model-driven beamformer (Nix & Hohmann, 2002)

- 2 Sources, unprocessed
- 2 Sources, processed, first direction // second direction
- 3 Sources, unprocessed//processed

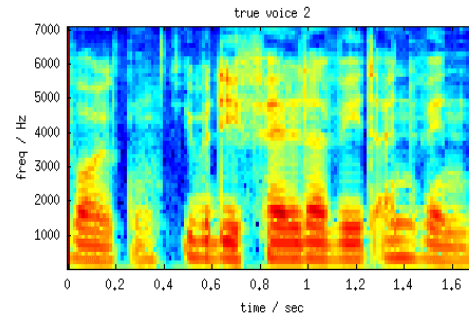
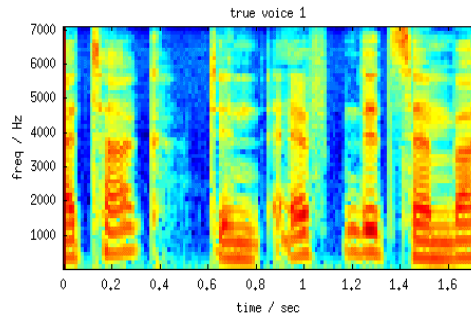


➔ No convincing separation of more than two sources

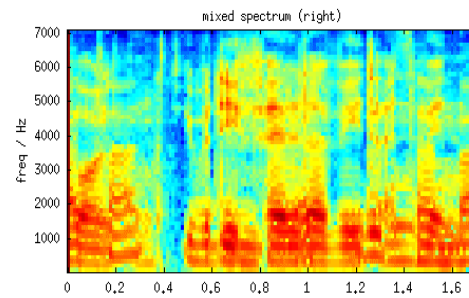
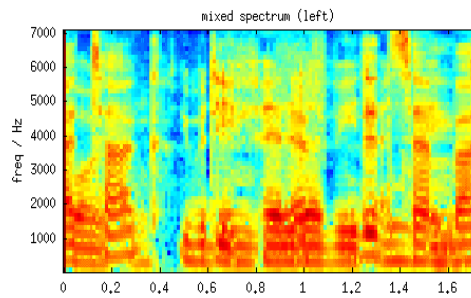
# Particle filter to estimate best beamformer online



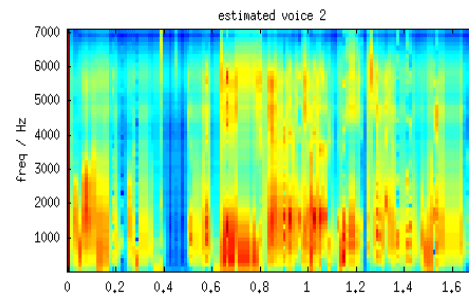
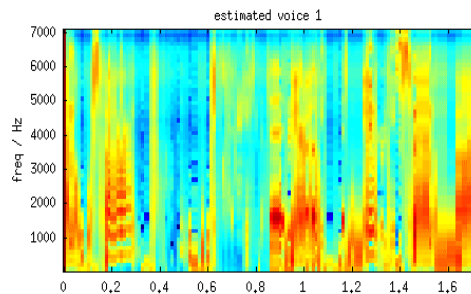
# Results: On-line Recovery of Spectral Envelopes



original voices



left/right ear  
signal

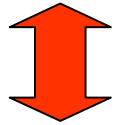


estimated  
voices (left side)

(see: Nix, Kleinschmidt, Hohmann, Tuesday 4pm)

➔ Separation from multiple sources with much computation & „cognitive“ complexity!

Test  
result



Predicted  
Test result



System  
perfor-  
mance

- Auditory principles for speech processing look promising
- Interaction experiment-model-application
- Amplitude Modulation Spectrogram !
- optimally switched two-sensor beamformer to mimic binaural system
- Top-down vs. bottom-up processing yet to be explored

# Hearing aid or personal communication device?



In-the-ear hearing aid



K-WON



HörTech Prototype



JABRA



Behind-the-ear  
hearing aid

Technology for hearing aids and mobile phones converge

→ Knowledge from hearing aid design is required for modern speech communication systems!







A binaural hearing aid to  
sit in –

*The ultimate way of  
achieving a good  
performance in cocktail  
parties!*

Auditory throne in  
front of the  
new „House of  
hearing“  
(Oldenburg)