

## Lecture 8: Spatial sound

- 1 Spatial acoustics
- 2 Binaural perception
- 3 Synthesizing spatial audio
- 4 Extracting spatial sounds

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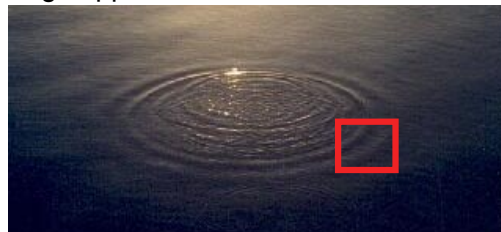
Columbia University Dept. of Electrical Engineering  
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### 1 Spatial acoustics

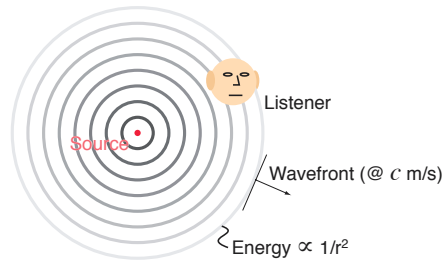
- **Received sound = source + channel**
  - so far, only considered ideal source waveform
- **Sound carries information on its spatial origin**
  - e.g. “ripples in the lake”



- evolutionary significance
- **The basis of scene analysis?**
  - yes and no - try blocking an ear



## Ripples in the lake

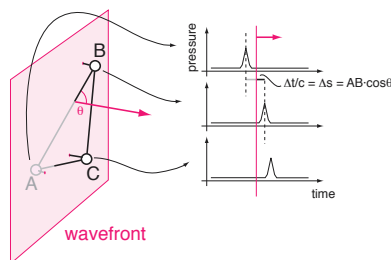


- **Effect of relative position on sound**
  - delay =  $\Delta r/c$
  - energy decay  $\sim 1/r^2$
  - absorption  $\sim G(f)^r$
  - direct energy plus reflections
- **Give cues for recovering source position**
- **Describe wavefront by its normal**

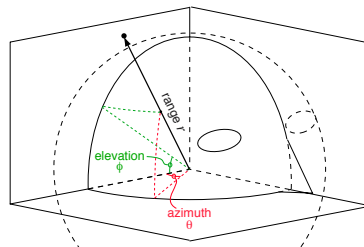


## Recovering spatial information

- **Source direction as wavefront normal**
  - moving plane found from timing at 3 points



- need to solve **correspondence**

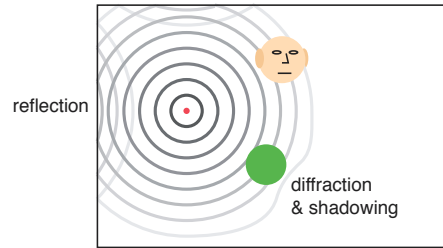


- **Space:**  
**need 3 parameters**
  - e.g. 2 angles and range



## The effect of the environment

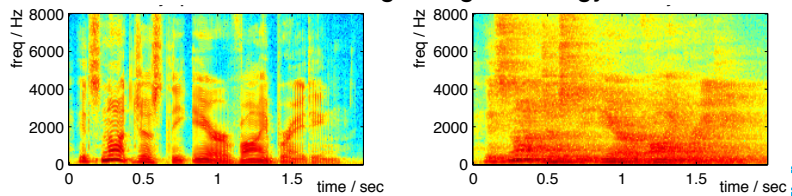
- **Reflection causes additional wavefronts**



- + scattering, absorption
- many paths → many echoes

- **Reverberant effect**

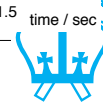
- causal 'smearing' of signal energy



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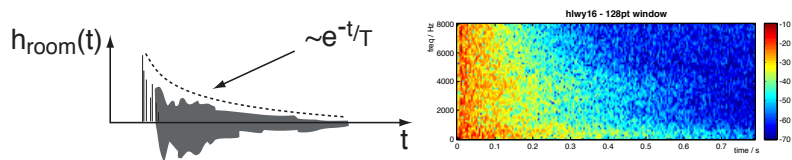
L08 - Spatial sound

2006-03-23 - 5



## Reverberation impulse response

- **Exponential decay of reflections:**



- **Frequency-dependent**
  - greater absorption at high frequencies  
→ faster decay
- **Size-dependent**
  - larger rooms → longer delays → slower decay

- **Sabine's equation:**

$$RT_{60} = \frac{0.049V}{S\bar{\alpha}}$$

- **Time constant as size, absorption**

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L08 - Spatial sound

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## Outline

- 1 Spatial acoustics
- 2 **Binaural perception**
  - The sound at the two ears
  - Available cues
  - Perceptual phenomena
- 3 Synthesizing spatial audio
- 4 Extracting spatial sounds

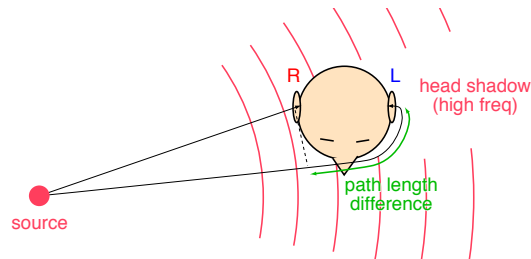


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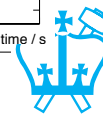
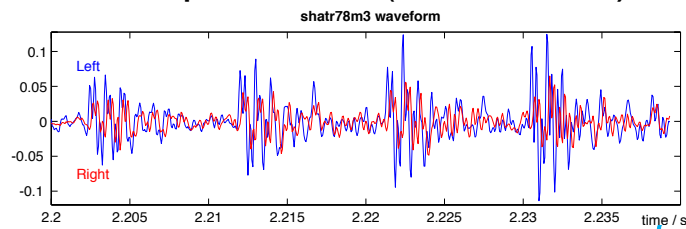
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## 2

## Binaural perception

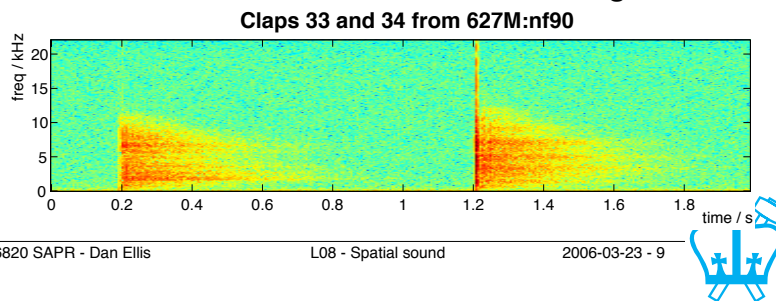


- **What is the information in the 2 ear signals?**
  - the **sound** of the source(s) (L+R)
  - the **position** of the source(s) (L-R)
- **Example waveforms (ShATR database)**



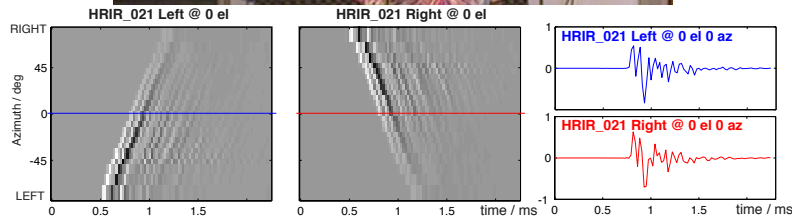
## Main cues to spatial hearing

- **Interaural time difference (ITD)**
  - from different path lengths around head
  - dominates in low frequency (< 1.5 kHz)
  - max ~ 750  $\mu$ s  $\rightarrow$  ambiguous for freqs > 600 Hz
- **Interaural intensity difference (IID)**
  - from head shadowing of far ear
  - negligible for LF; increases with frequency
- **Spectral detail (from pinna reflections) useful for elevation & range**
- **Direct-to-reverberant useful for range**



## Head-Related Transfer Fns (HRTFs)

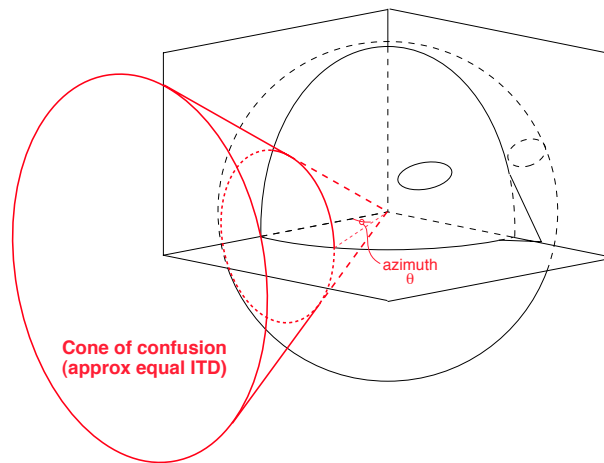
- Capture source coupling as impulse responses  
 $\{l_{\theta, \phi}, r_{\theta, \phi}, R(t)\}$
- Collection: (<http://interface.cipic.ucdavis.edu/>)



- **Highly individual!**



## Cone of confusion

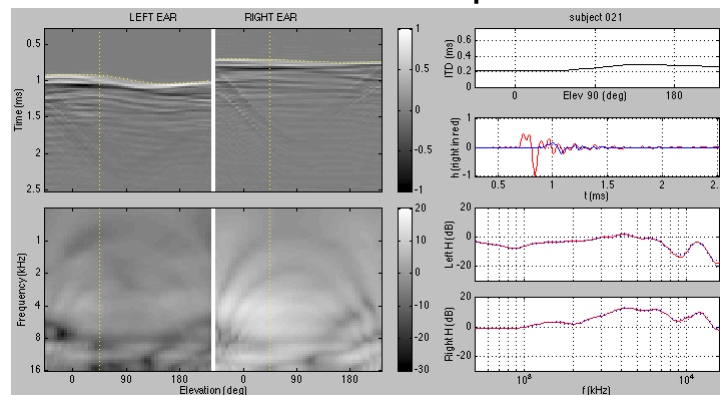


- **Interaural timing cue dominates (below 1kHz)**
  - from differing path lengths to two ears
- **But: only resolves to a cone**
  - Up/down? Front/back?



## Further cues

- **Pinna causes elevation-dependent coloration**



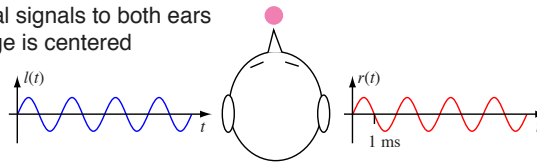
- **Monaural perception**
  - separate coloration from source spectrum?
- **Head motion**
  - synchronized spectral changes
  - also for ITD (front/back) etc.



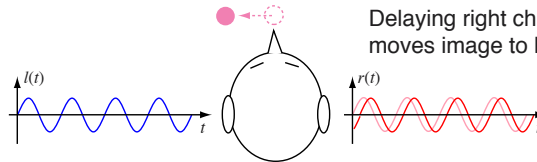
## Combining multiple cues

- Both **ITD** and **ILD** influence azimuth;  
What happens when they disagree?

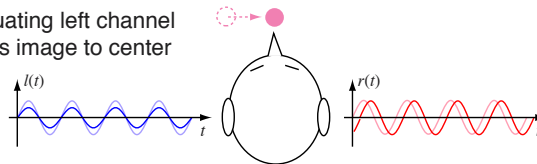
Identical signals to both ears  
→ image is centered



Delaying right channel  
moves image to left



Attenuating left channel  
returns image to center

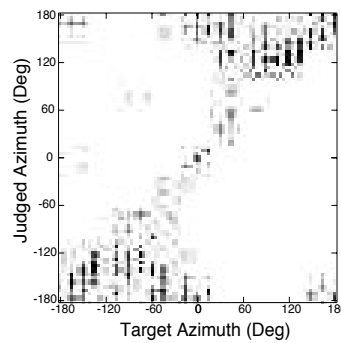


- trading @ around 0.1 ms / dB



## Binaural position estimation

- **Imperfect results:** (Arruda, Kistler & Wightman 1992)

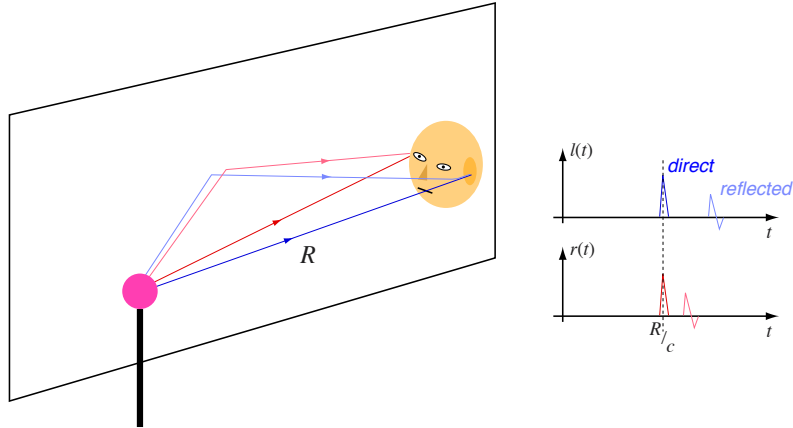


- listening to 'wrong' hrtfs → errors
- front/back reversals stay on cone of confusion



## The Precedence Effect

- **Reflections** give misleading spatial cues



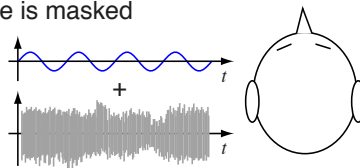
- **But: Spatial impression based on 1st wavefront** then 'switches off' for  $\sim 50$  ms
  - .. even if 'reflections' are louder
  - .. leads to impression of room



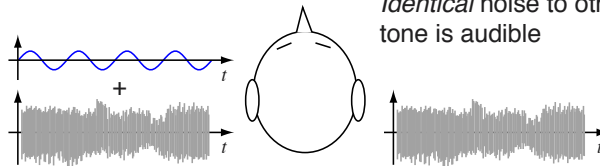
## Binaural Masking Release

- **Adding noise to reveal target**

Tone + noise to one ear:  
tone is masked



Identical noise to other ear:  
tone is audible



- why does this make sense?

- **Binaural Masking Level Difference up to 12dB**
  - greatest for noise in phase, tone anti-phase





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## Outline

- 1 Spatial acoustics
- 2 Binaural perception
- 3 Synthesizing spatial audio**
  - Position
  - Environment
- 4 Extracting spatial sounds



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## 3 Synthesizing spatial audio

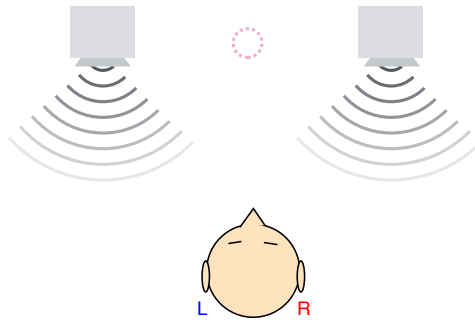
- **Goal: recreate realistic soundfield**
  - hi-fi experience
  - synthetic environments (VR)
- **Constraints**
  - resources
  - information (individual HRTFs)
  - delivery mechanism (headphones)
- **Source material types**
  - live recordings (actual soundfields)
  - synthetic (studio mixing, virtual environments)



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## Classic stereo



- **'Intensity panning':**  
**no timing modifications, just vary level  $\pm 20$  dB**
  - works as long as listener is equidistant (ILD)
- **Surround sound:**  
**extra channels in center, sides, ...**
  - same basic effect - pan between pairs

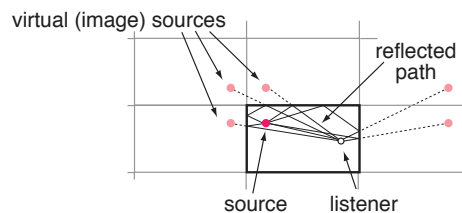


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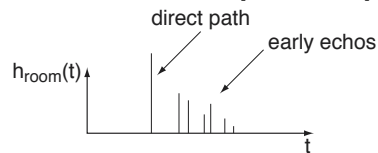
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## Simulating reverberation

- **Can characterize reverb by impulse response**
  - spatial cues are important - record in stereo
  - IRs of  $\sim 1$  sec  $\rightarrow$  **very** long convolution
- **Image model: reflections as duplicate sources**



- **'Early echos' in room impulse response:**



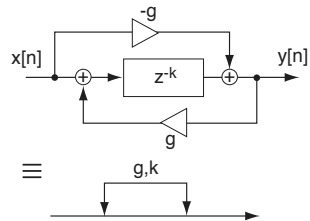
- **Actual reflection may be  $h_{\text{reflect}}(t)$ , not  $\delta(t)$**



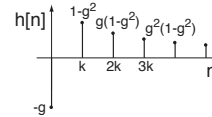
## Artificial reverberation

- **Reproduce perceptually salient aspects**
  - **early echo** pattern (→ room size impression)
  - overall decay tail (→ wall materials...)
  - interaural **coherence** (→ spaciousness)
- **Nested allpass filters** (Gardner '92)

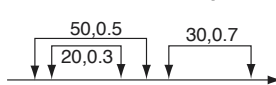
### Allpass



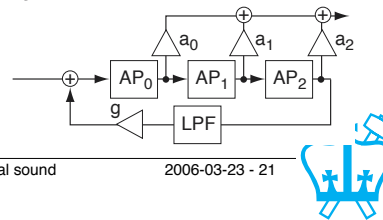
$$H(z) = \frac{z^{-k} - g}{1 - g \cdot z^{-k}}$$



### Nested+Cascade Allpass

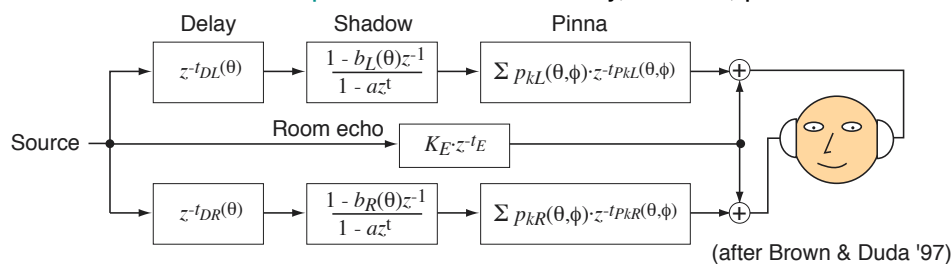


### Synthetic Reverb



## Synthetic binaural audio

- **Source convolved with {L,R} HRTFs gives precise positioning**
  - ...for headphone presentation
  - can combine multiple sources (by adding)
- **Where to get HRTFs?**
  - **measured set**, but: specific to individual, discrete
  - **interpolate** by linear crossfade, PCA basis set
  - or: **parametric model** - delay, shadow, pinna

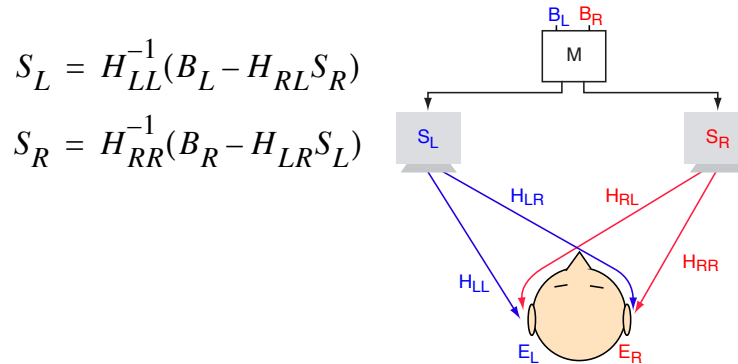


- **Head motion cues?**
  - head tracking + *fast* updates



## Transaural sound

- **Binaural** signals without headphones?
- Can **cross-cancel** wrap-around signals
  - speakers  $S_{L,R}$ , ears  $E_{L,R}$ , binaural signals  $B_{L,R}$ .

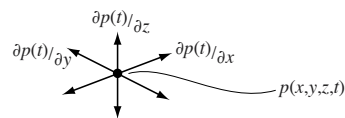


- **Narrow 'sweet spot'**
  - head motion?



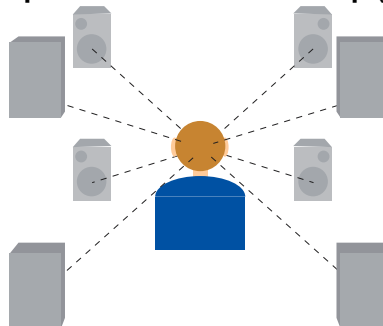
## Soundfield reconstruction

- **Stop thinking about ears**  
just reconstruct **pressure + spatial derivatives**



- ears in reconstructed field receive same sounds

- **Complex reconstruction setup (ambisonics)**



- able to preserve **head motion** cues?



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## Outline

- 1 Spatial acoustics
- 2 Binaural perception
- 3 Synthesizing spatial audio
- 4 **Extracting spatial sounds**
  - Microphone arrays
  - Modeling binaural processing



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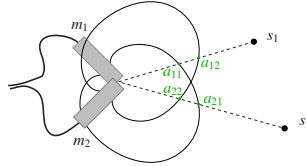
## 4 Extracting spatial sounds

- **Given access to soundfield, can we recover separate components?**
  - degrees of freedom:
    - >N signals from N sensors is hard
  - but: people can do it (somewhat)
- **Information-theoretic approach**
  - use only very general constraints
  - rely on precision measurements
- **Anthropic approach**
  - examine human perception
  - attempt to use same information



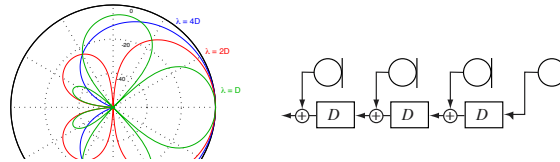
## Microphone arrays

- Signals from multiple microphones can be combined to enhance/cancel certain sources
- 'Coincident' mics with diff. directional gains



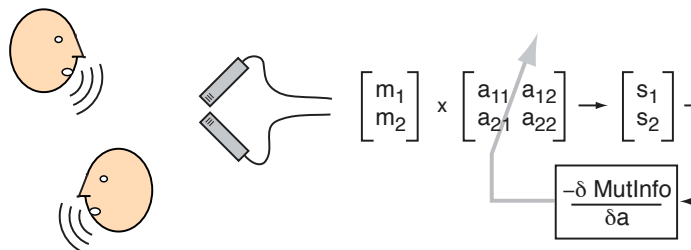
$$\begin{bmatrix} m_1 \\ m_2 \end{bmatrix} = \begin{bmatrix} a_{11} & a_{12} \\ a_{21} & a_{22} \end{bmatrix} \cdot \begin{bmatrix} s_1 \\ s_2 \end{bmatrix} \Rightarrow \begin{bmatrix} \hat{s}_1 \\ \hat{s}_2 \end{bmatrix} = \mathbf{A}^{-1} \cdot \mathbf{m}$$

- Microphone arrays (endfire)



## Adaptive Beamforming & Independent Component Analysis (ICA)

- Formulate mathematical criteria to optimize
- Beamforming: Drive interference to zero
  - cancel energy during nontarget intervals
- ICA: maximize mutual independence of outputs
  - from higher-order moments during overlap

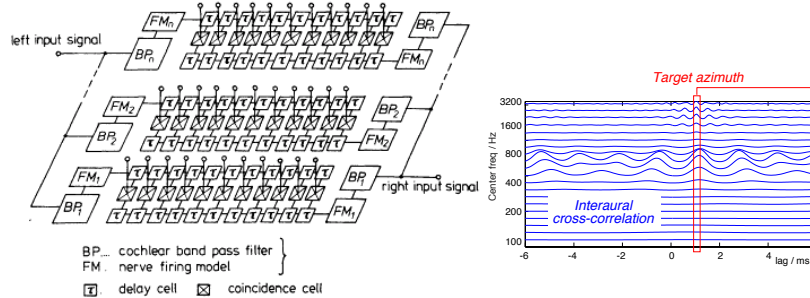


- Limited by separation model parameter space
  - only NxN?



## Binaural models

- **Human listeners do better?**
  - certainly given only 2 channels
- **Extract ITD and IID cues?**

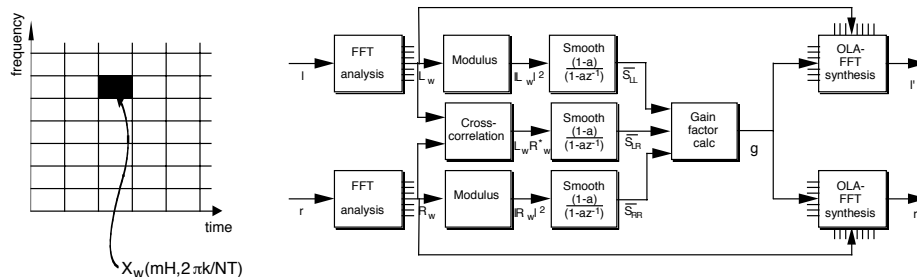


- cross-correlation finds timing differences
- 'consume' counter-moving pulses
- how to achieve IID, trading
- vertical cues...



## Nonlinear filtering

- **How to separate sounds based on direction?**
  - estimate direction locally
  - choose target direction
  - remove energy from other directions
- **E.g. Kollmeier, Peissig & Hohman '93**



- IID from  $|L_w|/|R_w|$ ; ITD (IPD) from  $\arg\{L_w R_w^*\}$
- match to IID/IPD template for desired direction
- also reverberation?



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## Summary

- **Spatial sound**
  - sampling at more than one point gives information on origin direction
- **Binaural perception**
  - time & intensity cues used between/within ears
- **Sound rendering**
  - conventional stereo
  - HRTF-based
- **Spatial analysis**
  - optimal linear techniques
  - elusive auditory models



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## References

- B.C.J. Moore, *An introduction to the psychology of hearing* (4th ed.) Academic, 1997.
- J. Blauert, *Spatial Hearing* (revised ed.), MIT Press, 1996.
- R.O. Duda, *Sound Localization Research*,  
<http://www.engr.sjsu.edu/~duda/Duda.Research.frameset.html>

