

## Mask-assisted speech enhancement for binaural hearing aids

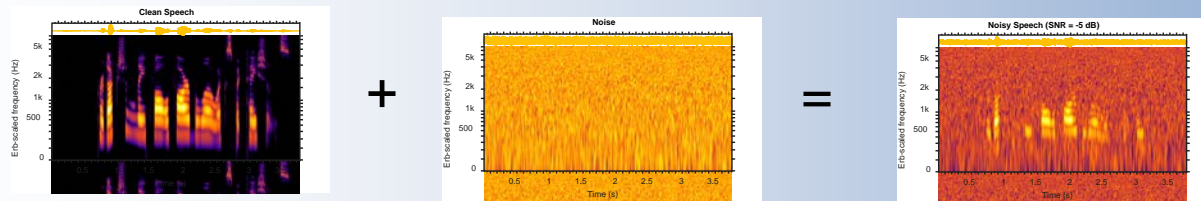
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Mike Brookes, Leo Lightburn, Alastair Moore,  
Patrick Naylor & Wei Xue

- Motivation: Ideal Binary Mask (IBM)
  - Intelligibility model for IBM-masked speech
  - STOI-optimal binary mask and its estimation
- Mask-assisted MMSE enhancement
  - Single-channel performance
- Binaural Enhancement
  - Alternatives for Metric reference signals
  - Bilateral versus Binaural beamforming
  - Effect of an improved mask
- Summary

## “Ideal” Binary Masks (IBM)

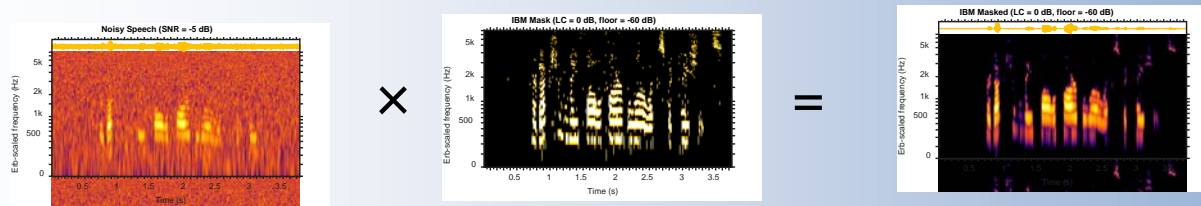
- Additive noise



SNR = -5 dB  
White Noise

- Apply Binary Mask

- Keep only time-frequency cells with local SNR > “local criterion” threshold (LC)



LC = 0 dB

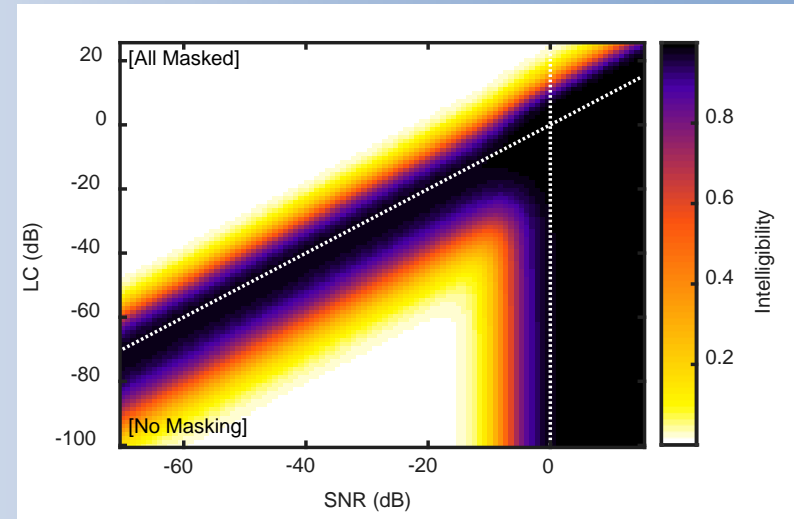
- An “oracle” mask has access to both the clean speech and the noise
  - In practice, the mask must be estimated from the noisy speech alone

# IBM-Masked Speech Intelligibility

# E-Lobes

∃ two independent sources of information: [Kjems et al 2010]

1. Noisy speech signal  
Distorted by the mask
2. Noise-vocoded signal  
Noise modulated by the mask

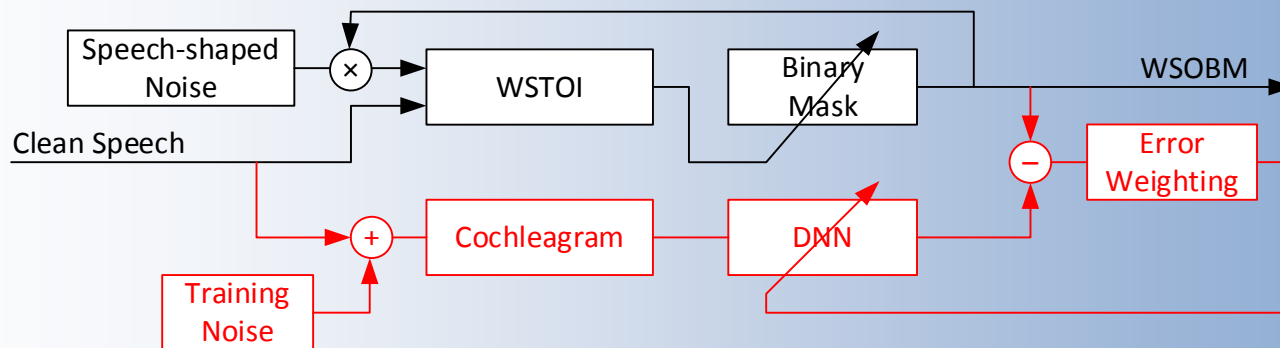


[from Kjems et al 2010]

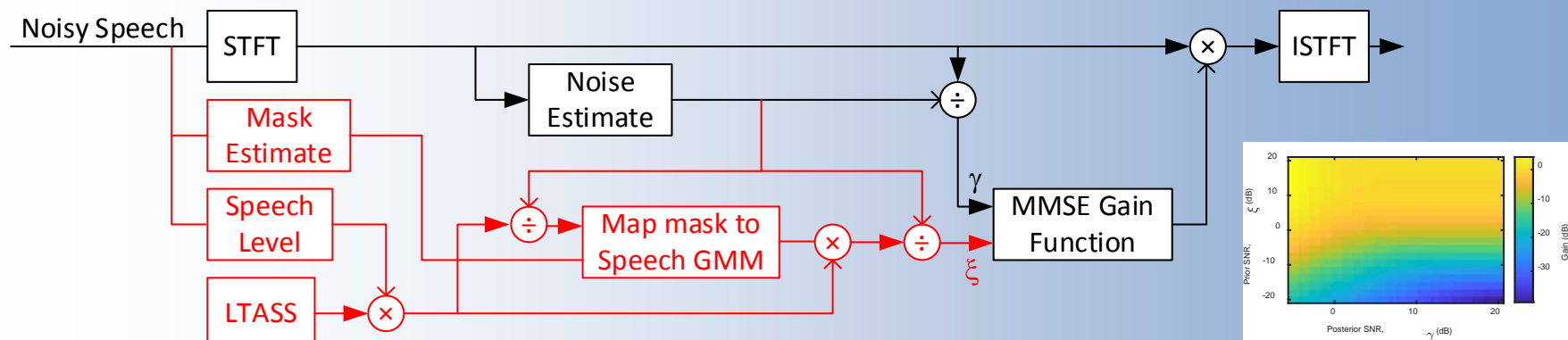
- Component 1 is intelligible for  $SNR > \approx -5$  dB provided mask is not too sparse ( $SNR > LC - 5$  dB)
  - vertical bar on figure
- Component 2 is intelligible if (a) high speech power  $\rightarrow$  mask on ( $SNR > LC - 5$  dB) and (b) low speech power  $\rightarrow$  mask off ( $SNR < LC + 20$  dB)
  - diagonal bar on figure

**(1) The benefit of binary masking comes entirely from component 2**

**(2) The mask should reflect clean speech energy (not the local SNR)**



- The STOI-optimal binary mask (SOBM) maximizes the STOI of masked speech-shaped noise (SSN)
  - Depends only on the clean speech
  - WSTOI weights time-frames by estimated speech information
- Train DNN to estimate the mask from noisy speech
  - Trained on a range of noises at a range of SNRs
  - Error weighting: (a) freq band importance, (b) WSTOI sensitivity
  - DNN output  $\in [0, 1]$  corresponds to probability that mask = 1

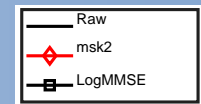
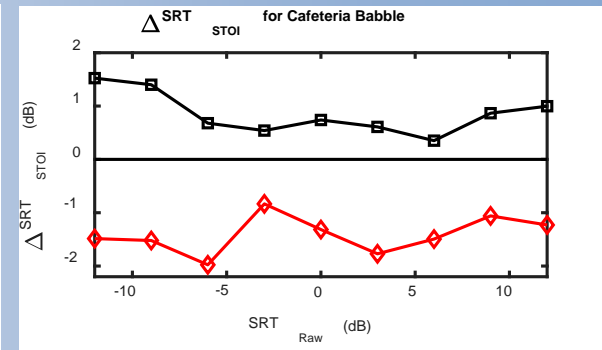
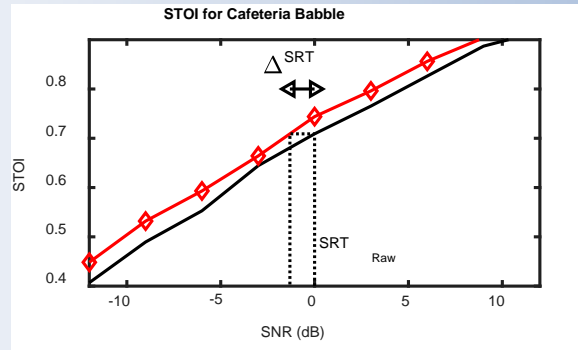


- LogMMSE enhancer assumes zero-mean complex Gaussian speech and noise STFT coefficient distributions
  - Gain function depends on posterior SNR,  $\gamma$ , and prior SNR,  $\xi$
- Map mask to Gaussian Mixture Model (GMM) distribution for speech power
  - Mapping depends on frequency band and estimated SNR
  - Denormalize by estimated speech level in the frequency band
  - Divide by estimated noise power to get GMM for prior SNR,  $\xi$

# Single-channel Enhancement

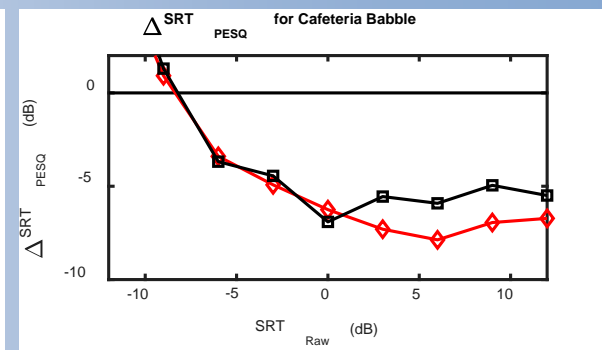
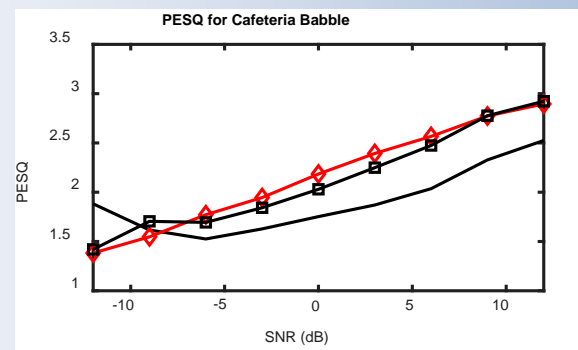
# E-Lobes

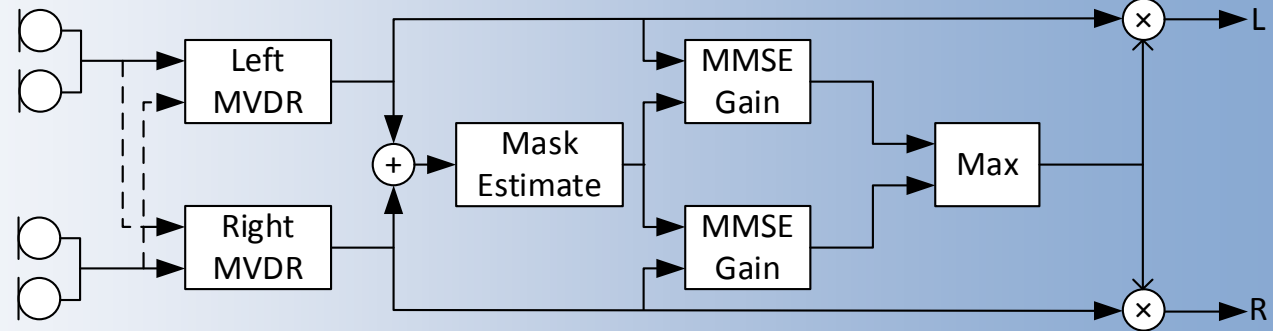
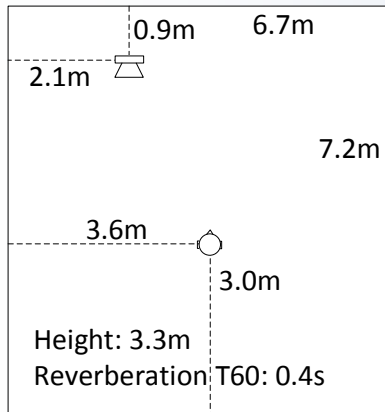
- Raw speech has acceptable intelligibility @  $\text{SNR} = \text{SRT}_{\text{Raw}}$
- Enhanced speech has the same intelligibility @  $\text{SRT}_{\text{Raw}} + \Delta\text{SRT}$



- Can regard  $-\Delta\text{SRT}$  as increased tolerance to noise
- Mask-assisted enhanced has  $\Delta\text{SRT}$  of  $-1.5$  dB
- In contrast, LogMMSE enhancer has  $\Delta\text{SRT}$  of  $+1$  dB

- PESQ tolerance to noise improves by  $>5$  dB for both enhancers at  $\text{SNR}_{\text{Raw}} > -5$  dB
  - Note: PESQ unreliable at low SNRs.





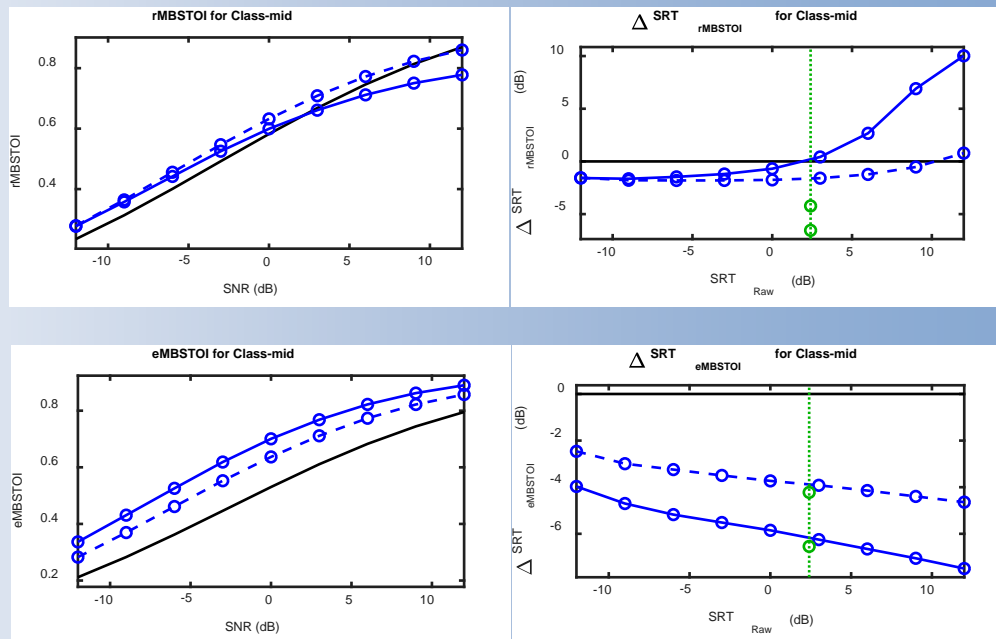
[Moore et al, 2018]

- Classroom full of noisy children. Highly non-stationary.
- Talker = loudspeaker, Listener = KEMAR head/torso simulator.
- MVDR beamformers:
  - Bilateral (2 mic): preserves spatial cues of noise sources
  - Binaural (4 mic): higher SNR, collapses noise to target direction
- Enhancement applies a time-frequency gain:
  - Common gain preserves binaural cues
  - Max function  $\approx$  “better ear”

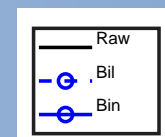


# Metric Reference Alternatives

- MBSTOI needs a clean speech reference:
  - Upper plots use **reverberant clean speech** as reference.
  - The green  $\circ$  shows the  $\Delta$  median-SRT @ 50% for 17 HI listeners.
  - Lower plots use the **early room response (50 ms)** to create the reference.



- When reverberant clean speech is used as the reference:
  - MBSTOI predicts small gains that do not match reality
  - Wrongly predicts that bilateral beamformer is better than binaural
- When early part of room response is used to create the reference:
  - MBSTOI correctly predicts  $\Delta$ SRT for both bilateral and binaural beamformers

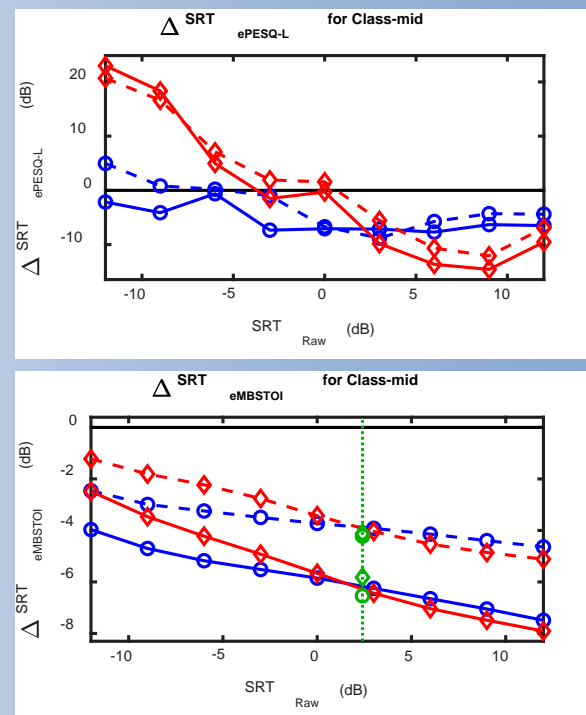


[MBSTOI: Andersen et al, 2018]

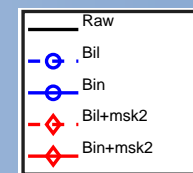
# Bilateral versus Binaural

- Binaural (solid lines) is always better than bilateral (dashed) for both PESQ and MBSTOI
- Enhancement,  $\blacklozenge$ , improves PESQ and MBSTOI for  $SRT_{Raw} > 2.5$  dB but degrades them below this.

– Worse than the single-channel results

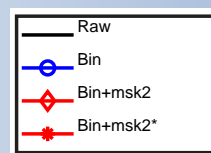
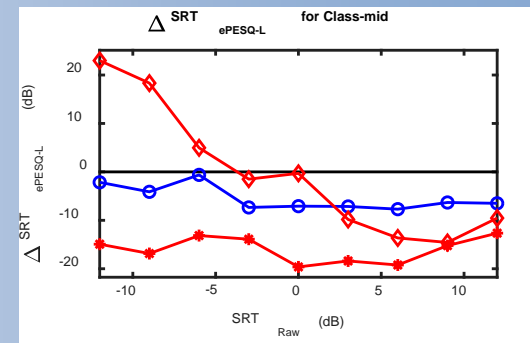
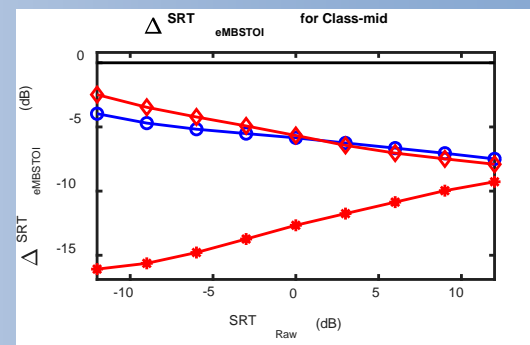
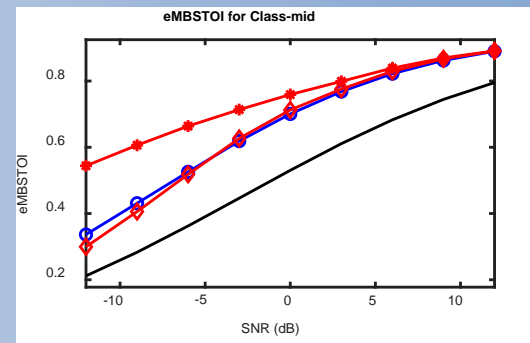


- Measured performance,  $\blacklozenge$ , of HI listeners shows that enhancement,  $\blacklozenge$ , degrades median SRT of binaural beamformer,  $\bullet$ , by 1 dB.



# Effect of Better Mask

- Effect of using a better mask (\* plot)
  - Fix the mask as the one determined for +12 dB SNR
  - MBSTOI declines more slowly with decreasing SNR
  - $\Delta SRT_{\text{MBSTOI}}$  continues to improve as SNR decreases
  - PESQ is improved at all SNRs
- Mask-assisted MMSE enhancement can give excellent results with a good enough mask



- Mask estimation
  - Aims to identify time-frequency cells that have high speech energy rather than high SNR (maximize STOI of vocoded noise)
  - Depends only on the target source and is single-channel
- Clean-speech reference for metrics
  - Metrics should use a non-reverberant clean-speech reference
  - Useful to express metric in terms of  $\Delta$ SRT
- Binaural versus Bilateral
  - For noise without dominant point sources, binaural  $\gg$  bilateral
  - Better SNR outweighs spatial cue preservation
- Mask-assisted LogMMSE enhancement
  - Can give significant gains but needs a better mask estimator

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