

# Background Adaptation for Improved Listening Experience in Broadcasting

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# Intelligibility issue in broadcast



- Factors causing low speech intelligibility [1]:
  - Background sound effects
  - Intrinsically unintelligible speech
  - Unfamiliar accents
  - Loud ambient noise
- Intelligibility enhancement [2-4] :
  - Reduced perceived quality of the modified speech [2]
  - Escalated annoyance when listening to modified speech.

[1] Armstrong et al, 2015 [2] Tang & Cooke, 2018; [3] Zoril et al, 2017; [4] Jokinen et al, 2016

# How about adapting the background sound(s)?



- Assumption
  - Both speech and background sound(s) are separately accessible (OBA).
  - Adapting the background sound may be less intrusive to listeners.
- Applying modification to the background signal
  - Can maintain the background level for design or artistic purposes
- Spectral weighting [1]
  - Similar to post-filtering: computationally cheap
  - Learning optimal weightings is time-consuming
  - Need a fast implementation for online processing

[1] Tang & Cooke, 2018

# Spectral weighting for background



- Adaptation: to reallocate the energy of the background,  $s$ , across 34 frequencies on the ERB scale.

$$s'(t) = k \cdot \sum_{f=1}^{F=34} s_f(t) \cdot 10^{W_f/20},$$

$s'$ : adapted  $s$

$k$ : scalar for renormalising the broadband signal energy

$W_f$ : spectral weighting

- Problem: to seek for a set of optimal  $W$

# Factors affecting overall listening experience



- Perceptual guides:
  - Speech intelligibility: High-Energy Glimpse proportion (HEGP [1-3])
  - Overall audio quality: Perceptual Evaluation of Audio Quality (PEAQ [4])
- A linear combination of HEGP and PEAQ
  - $k_{si}$ : weight for HEGP
  - $k_{aq}$ : weight for PEAQ

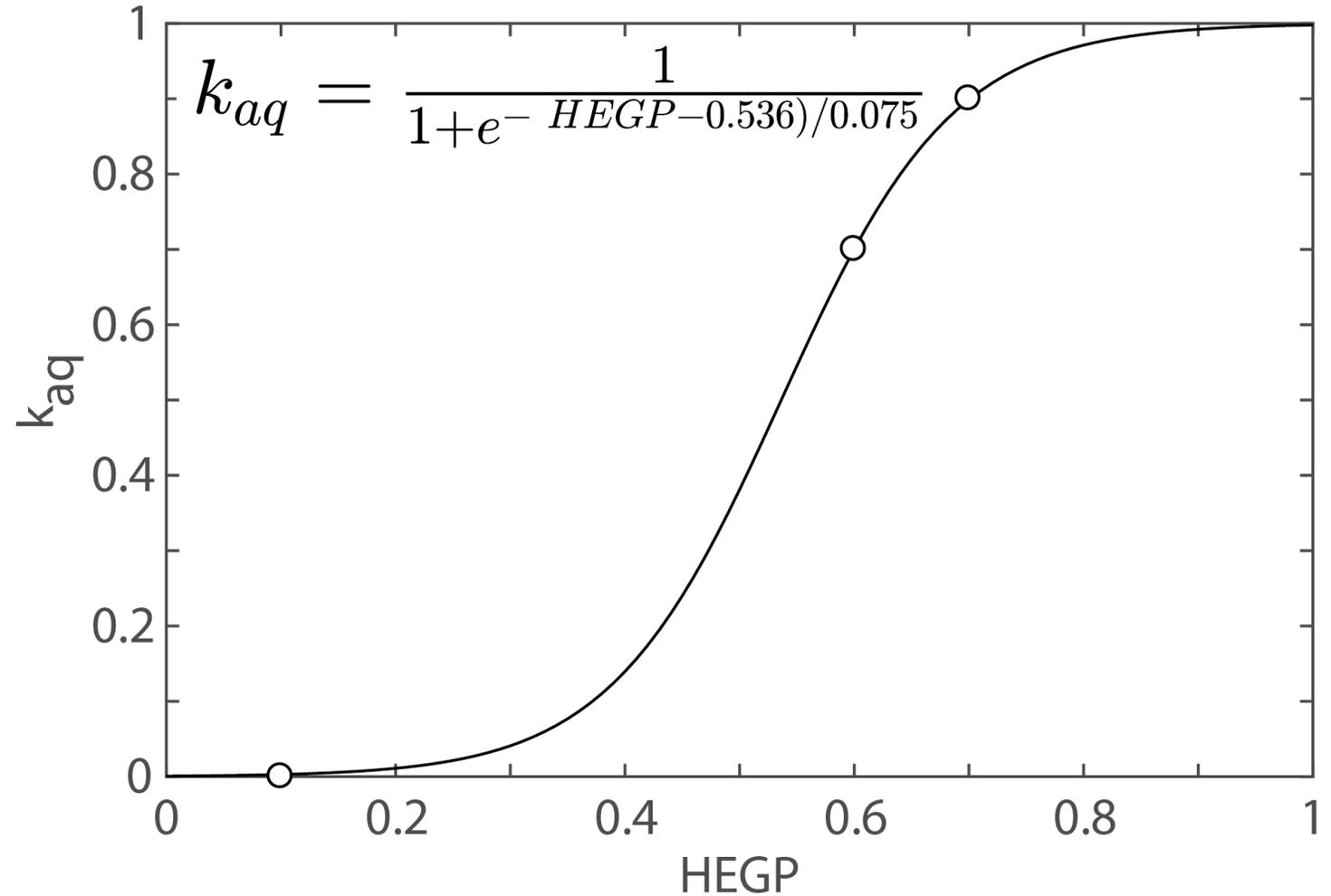
$$OM = k_{si} \cdot \text{HEGP} + k_{aq} \cdot \text{PEAQ}, \text{ w.r.t } k_{si} + k_{aq} = 1$$

[1] Tang & Cooke, 2016 [2] Tang et al, 2018a [3] Tang et al, 2018b [4] ITU-R BS.1387

# Relationship between intelligibility and audio quality



- When HEGP < 0.1, i.e. no intelligibility
  - Prioritising increasing intelligibility
  - $k_{aq} = 0$
- When HEGP  $\approx$  0.6, i.e. threshold of full intelligibility
  - Both intelligibility and quality affect listening experience
  - $k_{aq} = 0.7$
- When HEGP  $\approx$  0.7, i.e. more favourable SNR
  - Overall quality is dominant
  - $k_{aq} = 0.9$



# Closed-loop optimisation for spectral weightings $W$



- Task: to learn a set of optimal  $W_f$  (in dB) for each speech-background pair at a specified SNR.
- Optimisation procedure [1]
  - Algorithm: Pattern Search with MATLAB implementation
  - Variables: a vector of 34 elements, representing  $W_f$
  - Objective function: the linear combination of HEGP and PEAQ,  $OM$
- But Closed-loop optimisation is slow; not applicable for real time processing

[1] Tang et al, 2018a

# Neural network implementation



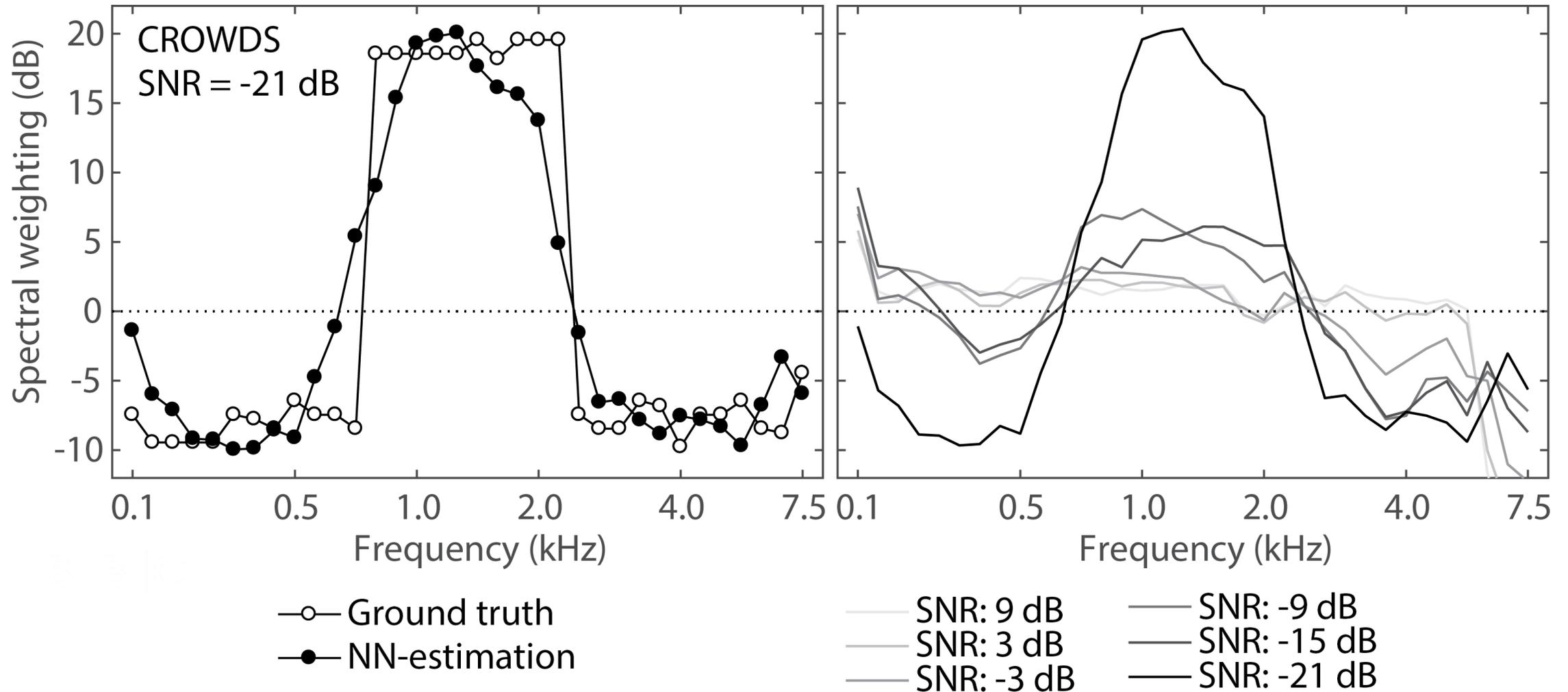
- A two-hidden-layer recurrent NN with backpropagation
- Input features:
  - 34 mean log-compressed speech spectra  $E_f^s$  and 34 noise spectra  $E_f^n$
  - 34 mean band SNRs, i.e.  $E_f^s - E_f^n$
  - A vector of 102 elements
- Grand-truth: 34 optimal weightings,  $W_f$ 
  - Learnt from maximising the linear combination of HEGP and PEAQ,  $OM$

# Experiments

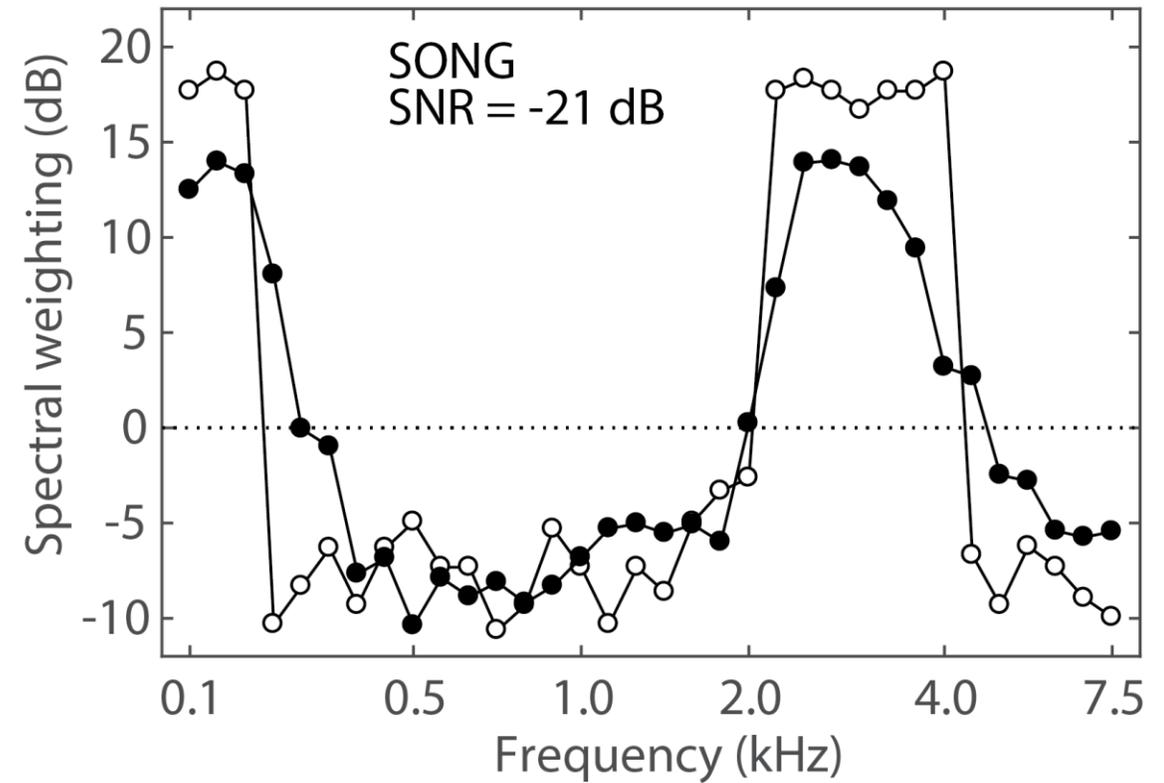
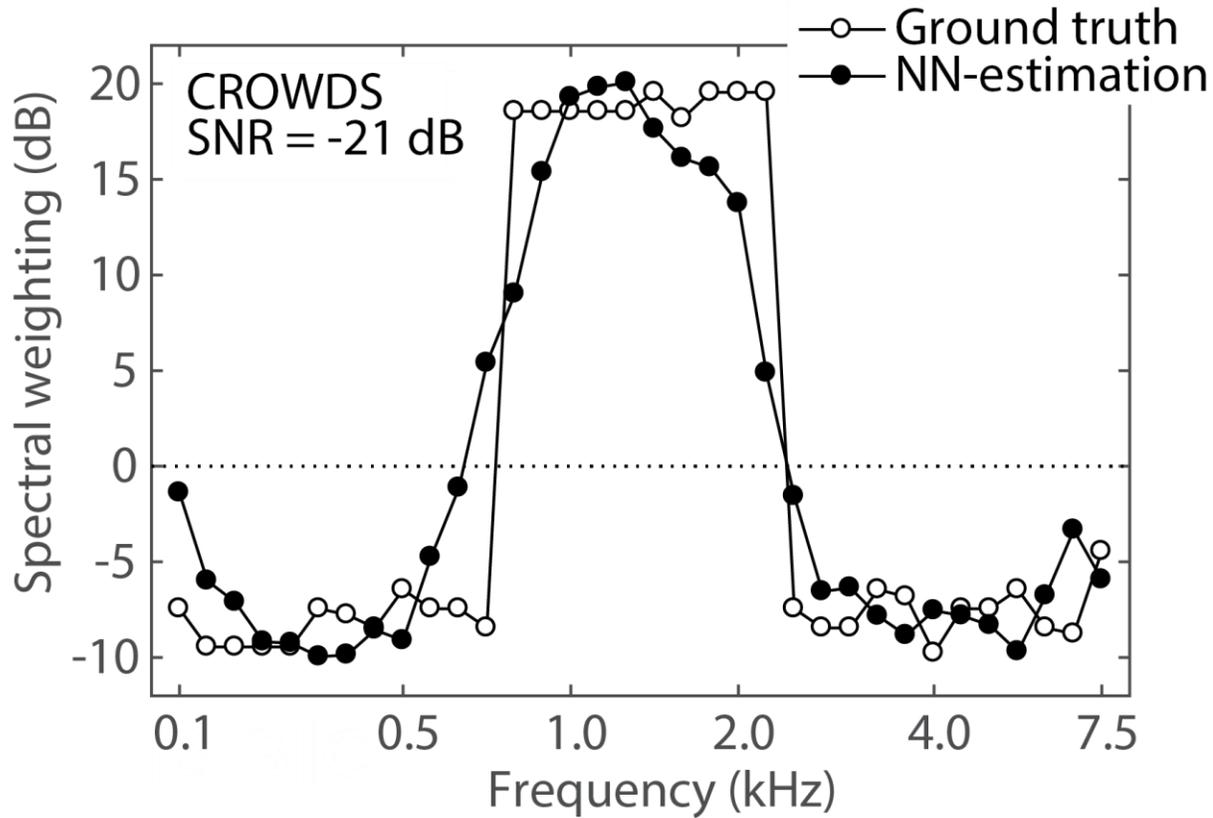


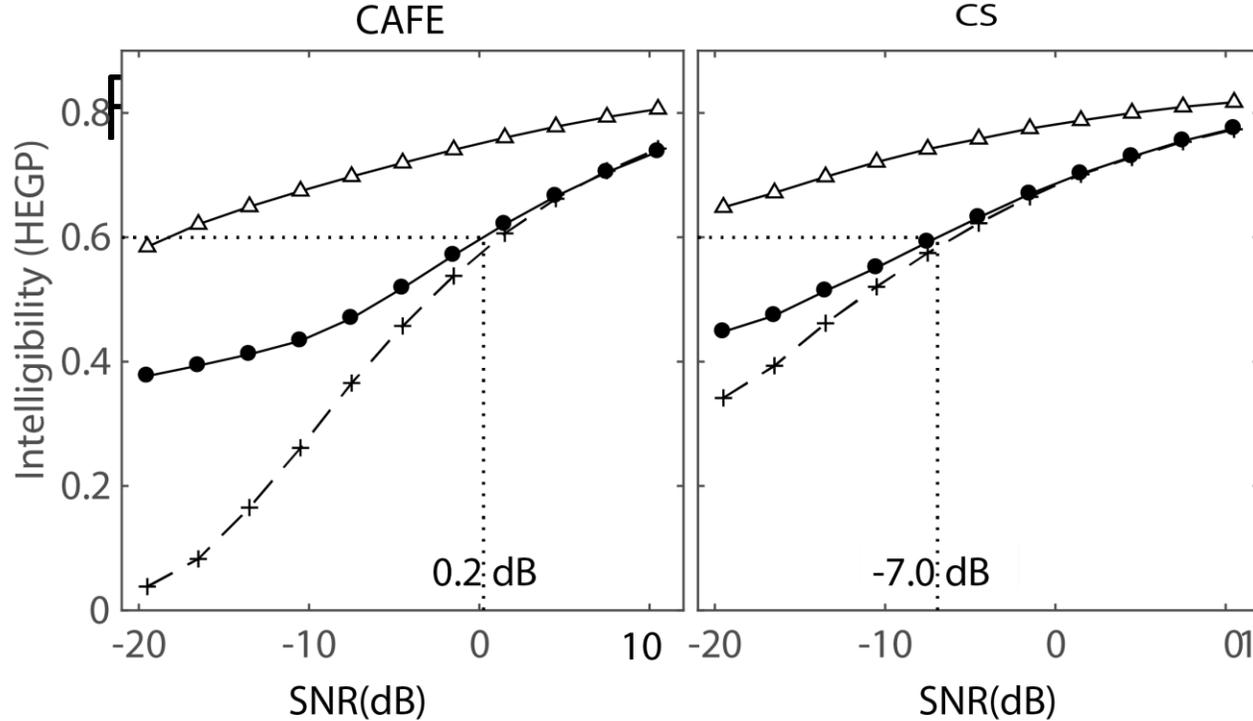
- NN Training data
  - 120 Harvard sentences sampled at 16 kHz; male talker
  - 6 background sounds:
    - ◆ café noise (CAFE)
    - ◆ female competing speech (CS)
    - ◆ stadium crowd noise (CROWDS)
    - ◆ a pop song (SONG)
    - ◆ the same song with vocal being removed (SONG-VR)
    - ◆ classic music (CLASSICAL)
  - SBRs: from -21 to 9 dB with steps of 3 dB
  - 7920 samples
- Test data
  - 300 sentences not appearing in training
  - SBRs: from -19.5 to 10.5 dB with steps of 3 dB

# Results I

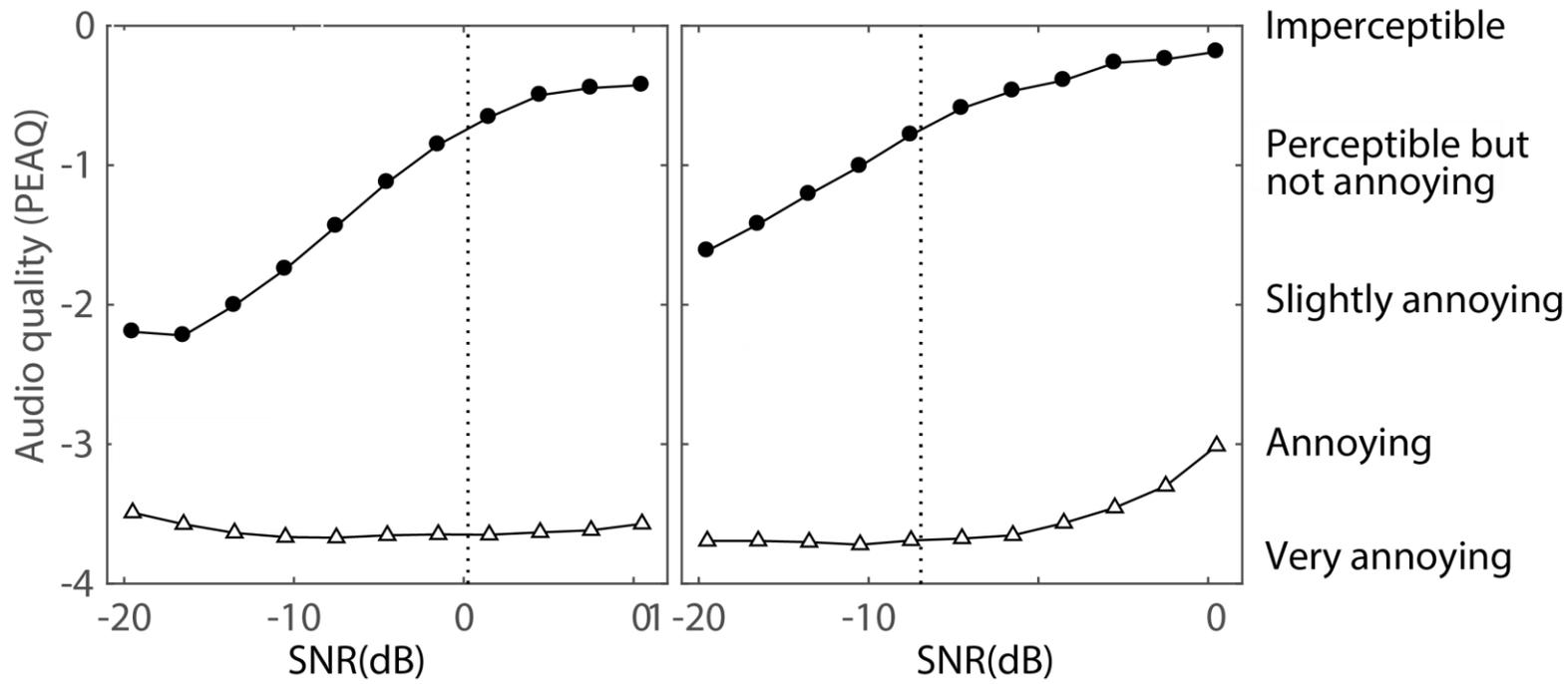


# Results II





- **Statically-weighted** leads to substantial HEGP gains at cost to the overall audio quality
- **Dynamically-weighted** shows more adaptive manner in preserving both intelligibility and audio quality



# Conclusions



- Spectral weighting inspired by near-end intelligibility enhancement is applied to the background signal, in order to enhance speech intelligibility while preserving the overall audio quality.
- With an adaptive function which models the relationship between intelligibility and audio quality, the optimised spectral weightings balance the two factors while modifying the background signal.
- A pre-trained NN is able to estimate the optimal spectral weightings from easy-to-compute acoustic features.
- Perceptual listening experiments are needed for further validating the method.

# Thank you!

This work was supported by the EPSRC Programme Grant S3A: Future Spatial Audio for an Immersive Listener Experience at Home (EP/L000539/1) and the BBC as part of the BBC Audio Research Partnership.