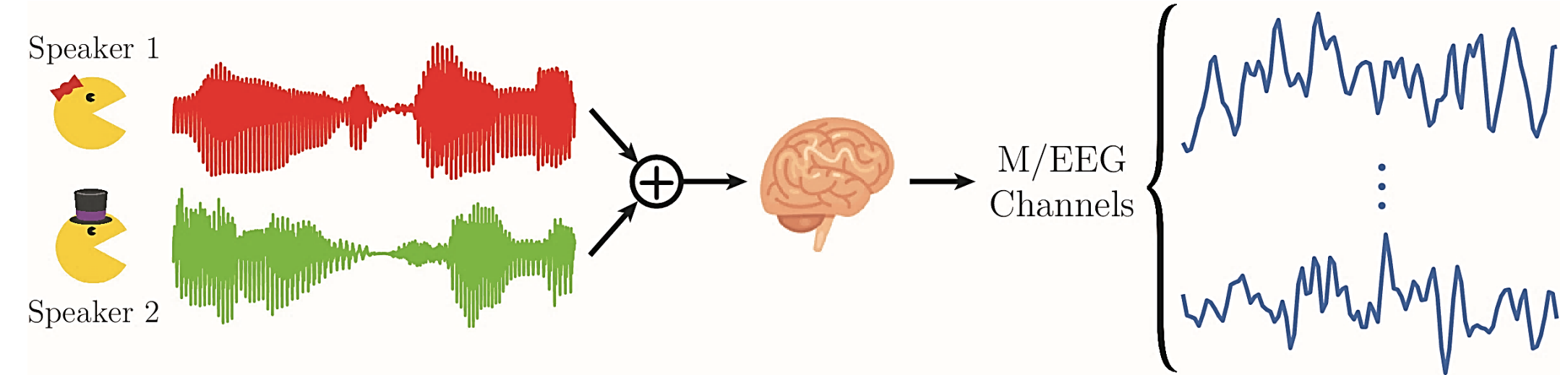


Problem Overview

Cocktail Party Effect: the ability to identify and track a target speaker amid a cacophony of acoustic interference [1]



Simplified Computational Problem: In a *dual-speaker* environment, can we decode the attentional state in *real-time* from the *clean speech signals* of the two speakers and the multi-channel *magnetoencephalography (MEG)* or *electroencephalography (EEG)* measurements of the listener's brain?

Applications: brain-computer Interface (BCI) systems and smart hearing aids

decoding models \rightarrow linearly map M/EEG data to stimulus
encoding models \rightarrow linearly map stimulus to a neural response from M/EEG

Existing Methods:

- reverse-correlation or stimulus reconstruction in decoding models (EEG) [2]:** train a decoder on the *attended* speech using training data; apply the trained decoder on recorded EEG to reconstruct a stimulus; speech that best matches the reconstruction is classified as the attended speech
- important stimulus time lags in encoding models (MEG) [3][4]:** estimate the encoding coefficients for each speaker, i.e., Temporal Response Function (TRF); the speaker with a larger M100 (the TRF peak close to 100ms delay) is classified as the attended speaker

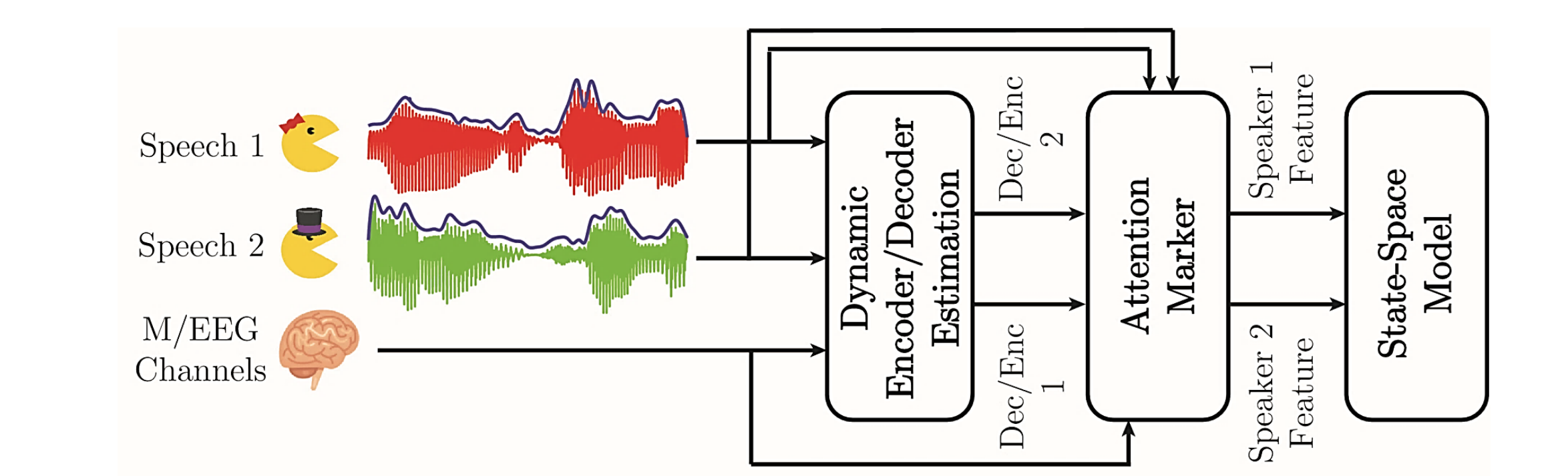
Shortcomings for Real-Time Attention Decoding:

- attention decoding accuracy drops significantly at high temporal resolutions, e.g. 1s (**unreliable performance in real-time settings**)
- need *large training datasets* to pre-estimate the *attended* encoder/decoder coefficients reliably (**may not be accessible in real-time applications**)

References

- [1] Cherry, E. Colin. "Some experiments on the recognition of speech, with one and with two ears." *The Journal of the acoustical society of America* 25.5 (1953): 975-979.
- [2] O'Sullivan, James A., et al. "Attentional selection in a cocktail party environment can be decoded from single-trial EEG." *Cerebral Cortex* 25.7 (2014): 1697-1706.
- [3] Ding, Nai, and Jonathan Z. Simon. "Emergence of neural encoding of auditory objects while listening to competing speakers." *Proceedings of the National Academy of Sciences* 109.29 (2012): 11854-11859.
- [4] Akram, Sahar, Jonathan Z. Simon, and Behtash Babadi. "Dynamic Estimation of the Auditory Temporal Response Function From MEG in Competing-Speaker Environments." *IEEE Transactions on Biomedical Engineering* 64.8 (2017): 1896-1905.
- [5] Akram, Sahar, et al. "Robust decoding of selective auditory attention from MEG in a competing-speaker environment via state-space modeling." *NeuroImage* 124 (2016): 906-917.

Proposed Framework



Dynamic Encoder/Decoder Estimation:

- consider K consecutive non-overlapping windows of length W samples
- update the enc./dec. estimates $\hat{\theta}_k^{(i)}$ for *each speaker* in *every window*:

$$\hat{\theta}_k^{(i)} = \arg \min_{\theta} \sum_{j=1}^k \lambda^{k-j} \|y_j - X_j \theta\|_2^2 + \gamma \|\theta\|_1, \quad k = 1, 2, \dots, K, \quad i = 1, 2$$

forgetting factor \leftarrow ℓ_1 regularization penalty

speech envelopes (dec.)
neural response (enc.)

M/EEG covariates (dec.)
envelope covariates (enc.)

Attention Marker:

- an *attention-modulated* feature for *each speaker* in *every window*

$$m_k^{(i)} = f(y_k, X_j, \hat{\theta}_k^{(i)})$$
- potential examples:**
 - reverse-correlation in dec. models: $m_k^{(i)} = |\text{corr}(y_k^{(i)}, X_j \hat{\theta}_k^{(i)})|$
 - M100 peak magnitude in MEG enc. models: $|\hat{\theta}_k^{(i)}|$ near the 100ms delay

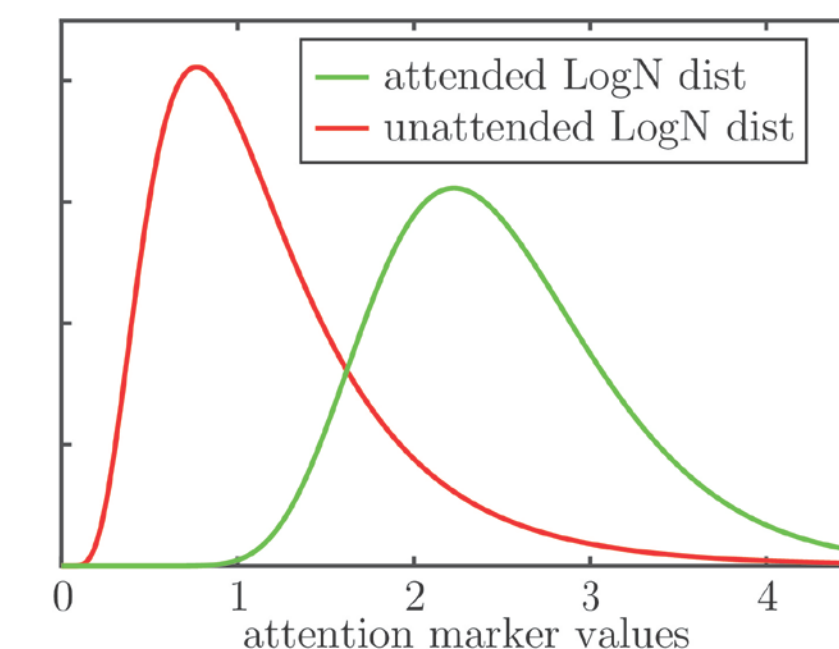
Dynamic State-Space Model:

- $n_k = 1$ (resp. 2) if speaker 1 (resp. 2) is attended at window k
- fit the following model on $m_k^{(i)}$'s in a *fixed-lag sliding window* fashion for real-time attention decoding

$$\begin{cases} m_k^{(i)} | n_k = i \sim \text{LogNormal}(\rho^{(a)}, \mu^{(a)}) \\ m_k^{(i)} | n_k \neq i \sim \text{LogNormal}(\rho^{(u)}, \mu^{(u)}) \end{cases}$$

$$\begin{cases} p_k = P(n_k = 1) = \frac{1}{1 + \exp(-z_k)} \\ z_k = z_{k-1} + w_k \\ w_k \sim N(0, \eta_k) \end{cases}$$

- model parameters:** z_k 's, η_k 's, $\rho^{(a)}$, $\rho^{(u)}$, $\mu^{(a)}$, $\mu^{(u)}$
- goal at window $k = k_0$:** estimate $p_{k^*} = \text{logistic}(z_{k^*})$ where $k^* = k_0 - K_F$ through nested EMs [4] as a *dynamic*, *probabilistic*, and *robust* measure of the attentional state



EEG Analysis (Decoding Model)

Experiment Specifications:

- 3 subjects, *instructed* constant attention on speaker 1, two speakers
- 64-channel EEG recording, 24 trials each 60s
- EEG signals and speech envelopes downsampled to $f_s = 64\text{Hz}$

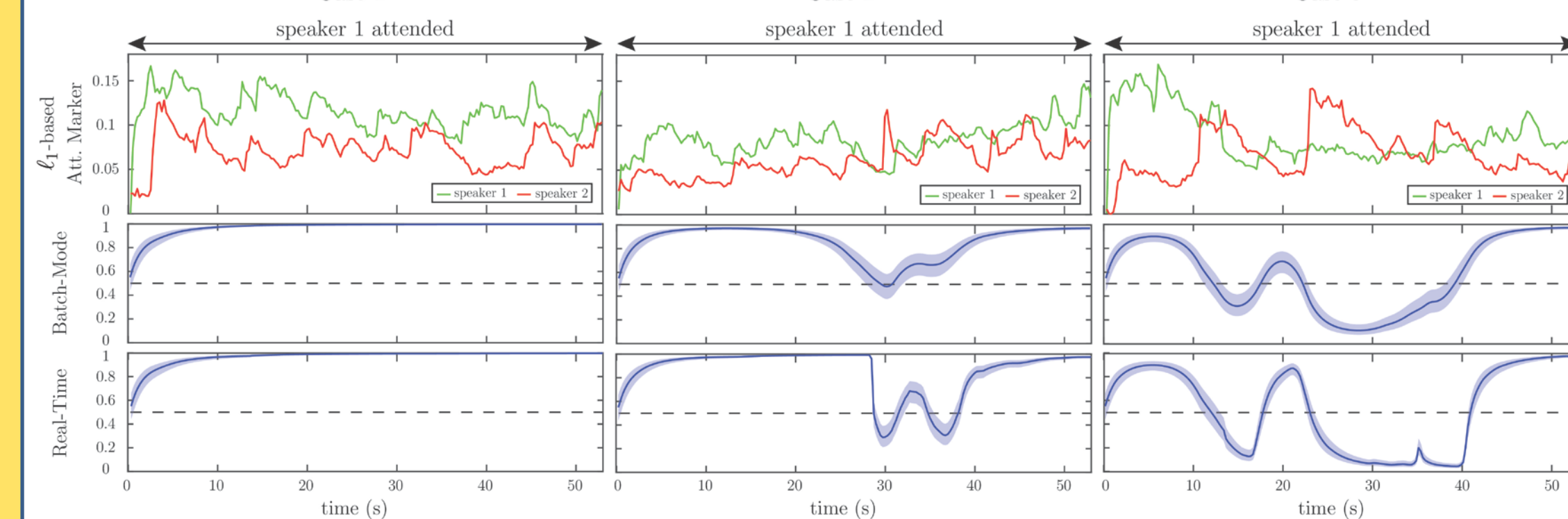
Attention Decoding Framework:

- decoder estimation parameters:** $W = 0.25f_s$, decoder lag of 0.25s
- attention marker:** ℓ_1 norm of the decoder, i.e., $m_k^{(i)} = \|\hat{\theta}_k^{(i)}\|_1$
 \leftarrow **rationale:** detects significant peaks in dec. coefficients
- total built-in attention decoding delay:** 1.5s + 0.25s = 1.75s

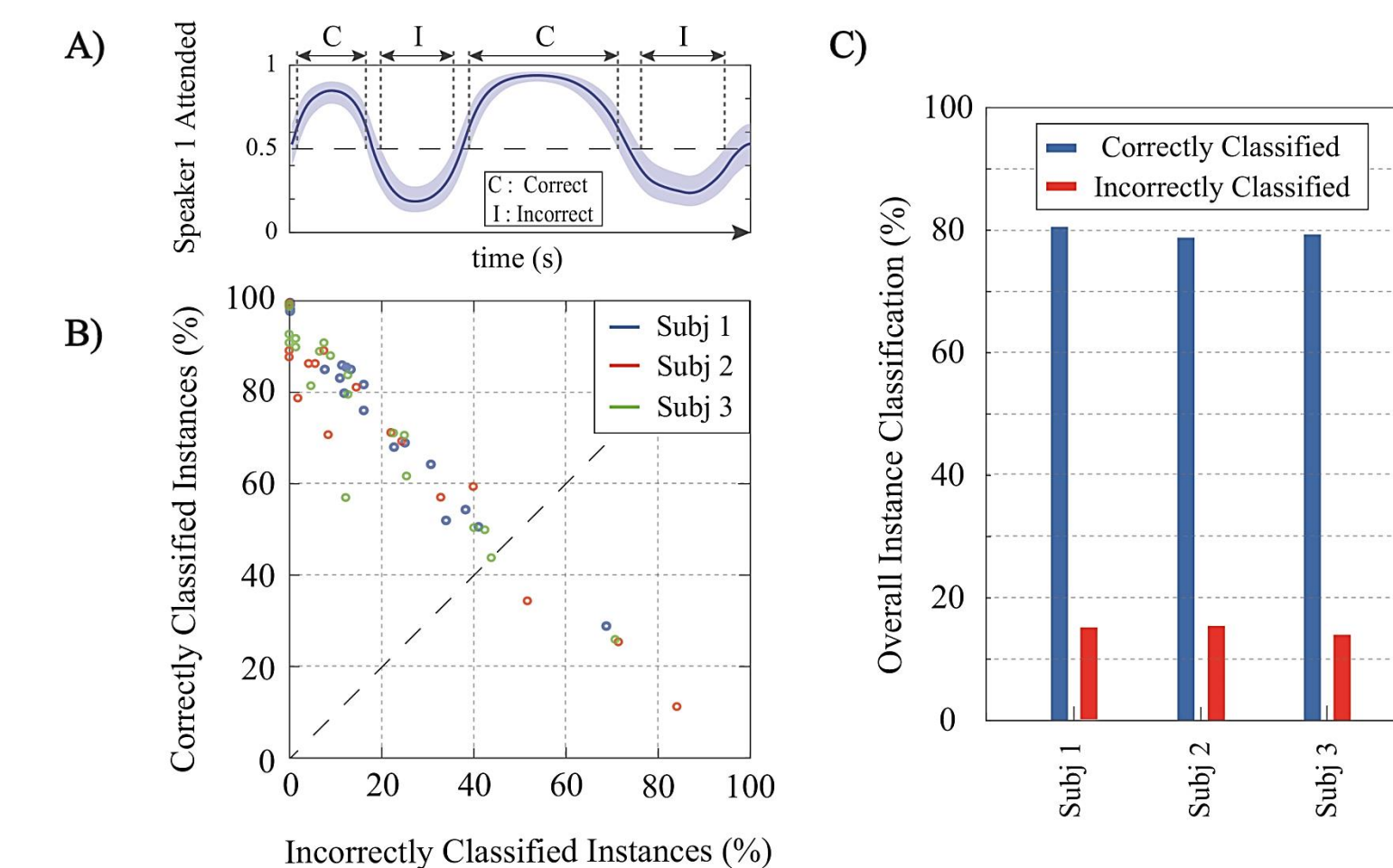
forward-lag in state-space model application \leftarrow decoder lag

Example Trial Outputs:

- separating power of the attention marker decreasing from case 1 to 3
- second row shows inferred p_k 's in our real-time framework
- third row shows inferred p_k 's in the batch-mode case, where the state-space processes all $m_k^{(i)}$'s at once

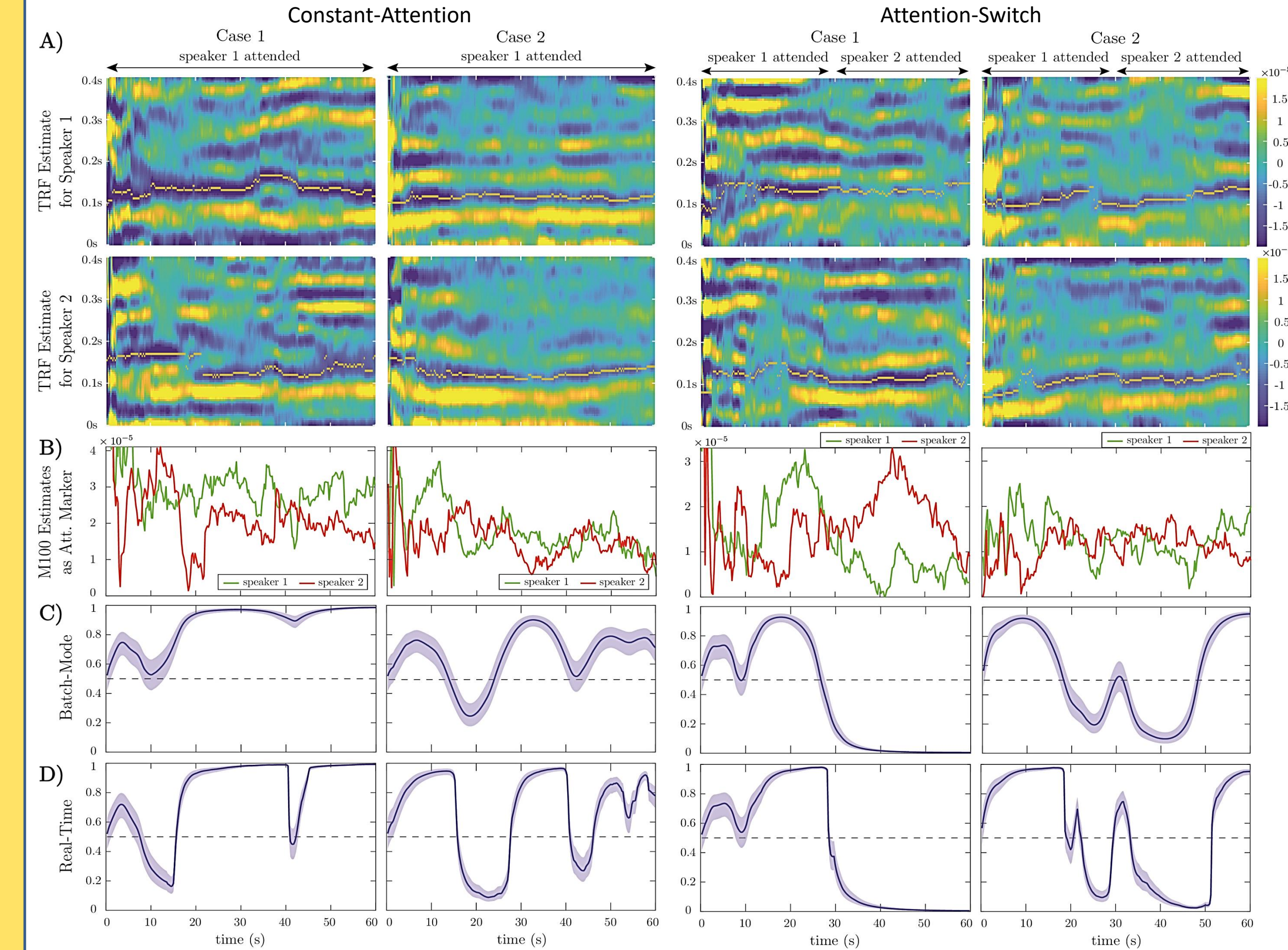


average classification accuracy in a trial for each subject:

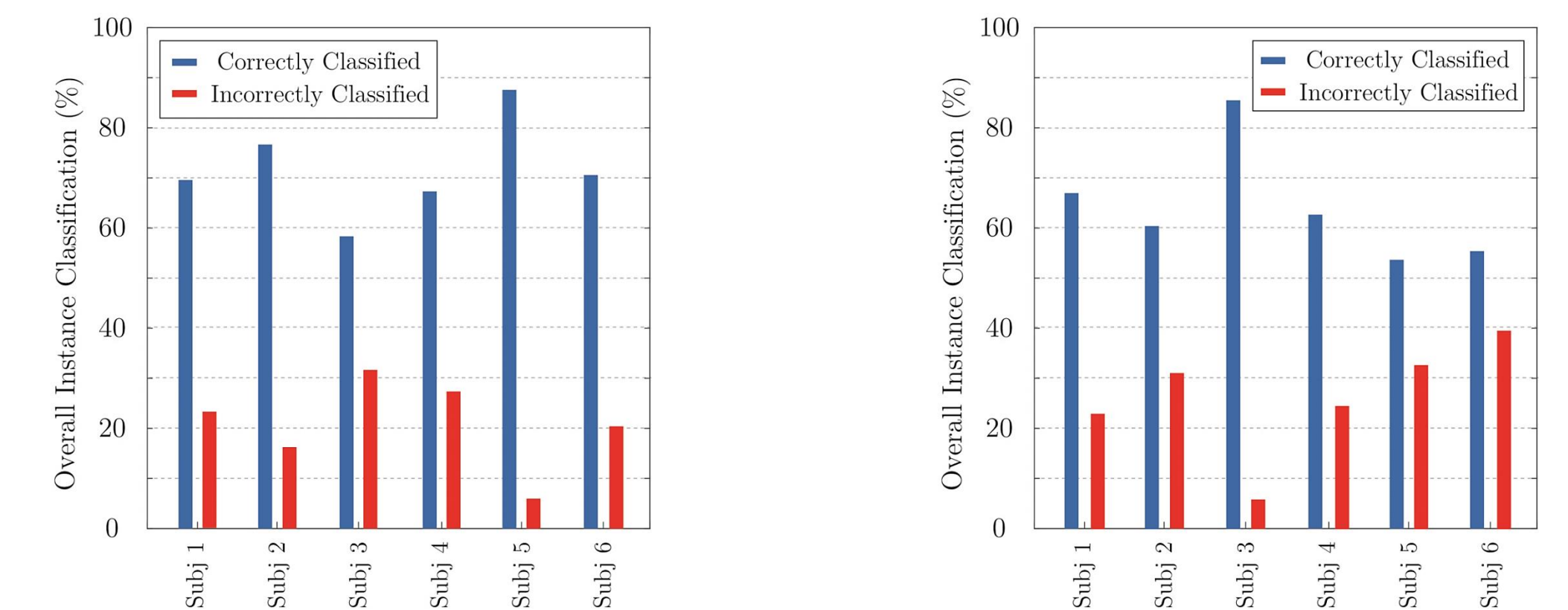


MEG Analysis (Encoding Model)

- 6 subjects, two speakers, constant-attention (6 trials) and attention-switch (3 trials) experiments
- attention marker:** real-time M100 magnitude estimates in the TRFs
example TRF estimation results and state-space outputs:



average classification accuracy in a trial for each subject:



Summary

- a new framework for real-time attention decoding in competing speaker settings
 - \leftarrow *real-time* estimation of encoding or decoding coefficients
 - \leftarrow computing an att.-modulated feature from the estimates and recorded data
 - \leftarrow apply a state-space model on the features for a *statistically interpretable* and *robust* measure of the attentional state
- can operate at high temporal resolution with no need for large training datasets, unlike existing methods