Cortical Representations of Speech in a Multi-talker Auditory Scene

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Abstract

The ability to parse a complex auditory scene into perceptual objects is facilitated

by a hierarchical auditory system. Successive stages in the hierarchy transform an

auditory scene of multiple overlapping sources, from peripheral tonotopically-

based representations in the auditory nerve, into perceptually distinct auditory-

objects based representation in auditory cortex. Here, using magnetoencephalo-

graphy (MEG) recordings from human subjects, we investigate how a complex

acoustic scene consisting of multiple speech sources is represented in distinct

hierarchical stages of auditory cortex. Using systems-theoretic methods of

stimulus reconstruction, we show that the primary-like areas in auditory cortex

contain dominantly spectro-temporal based representations of the entire auditory

scene. Here, both attended and ignored speech streams are represented with almost

equal fidelity, and a global representation of the full auditory scene with all its

streams is a better candidate neural representation than that of individual streams

being represented separately. In contrast, we also show that higher order auditory

cortical areas represent the attended stream separately, and with significantly

higher fidelity, than unattended streams. Furthermore, the unattended background

streams are more faithfully represented as a single unsegregated background

object rather than as separated objects. Taken together, these findings demonstrate

the progression of the representations and processing of a complex acoustic scene

up through the hierarchy of human auditory cortex.

Significance Statement:

Using magnetoencephalography (MEG) recordings from human listeners in a

simulated cocktail party environment, we investigate how a complex acoustic

scene consisting of multiple speech sources is represented in separate hierarchical

stages of auditory cortex. We show that the primary-like areas in auditory cortex

use a dominantly spectro-temporal based representation of the entire auditory

scene, with both attended and ignored speech streams represented with almost

equal fidelity. In contrast, we show that higher order auditory cortical areas

represent an attended speech stream separately from, and with significantly higher

fidelity than, unattended speech streams. Furthermore, the unattended background

streams are represented as a single undivided background object rather than as

distinct background objects.

Introduction

Individual sounds originating from multiple sources in a complex auditory scene mix linearly and irreversibly before they enter the ear, yet are perceived as distinct objects by the listener (Cherry, 1953; Bregman, 1994; McDermott, 2009). The separation, or rather individual re-creation, of such linearly mixed original sound sources is a mathematically ill-posed question, yet the brain nevertheless routinely performs this task with ease. The neural mechanisms by which this perceptual 'un-mixing' of sounds occur, the collective cortical representations of the auditory scene and its constituents, and the role of attention in both, are key problems in contemporary auditory neuroscience.

It is known that auditory processing in primate cortex is hierarchical (Davis and Johnsrude, 2003; Hickok and Poeppel, 2007; Rauschecker and Scott, 2009; Okada et al., 2010; Peelle et al., 2010; Overath et al., 2015) with subcortical areas projecting onto the core areas of auditory cortex, and from there, on to belt, parabelt and additional auditory areas (Kaas and Hackett, 2000). Sound entering the ear reaches different anatomical/functional areas of auditory cortex with different latencies (Recanzone et al., 2000; Nourski et al., 2014). Due to this serial component of auditory processing, the hierarchy of processing can be described by both anatomy and latency, of which the latter may be exploited using the high

temporal fidelity of non-invasive magnetoencephalography (MEG) neural

recordings.

In selective listening experiments using natural speech and MEG, the two

major neural responses known to track the speech envelope are the $M50_{TRF}$ and

$M100_{TRF}$, with respective latencies of $30-80$ ms and $80-150$ ms, of which the

dominant neural sources are, respectively, Heschl's gyrus (HG) and Planum

temporale (PT) (Steinschneider et al., 2011; Ding and Simon, 2012b).

Posteromedial HG is the site of core auditory cortex; PT contains both belt and

parabelt auditory areas (here collectively referred to as higher-order areas)

(Griffiths and Warren, 2002; Sweet et al., 2005). Hence the earlier neural

responses are dominated by core auditory cortex, and the later are dominated by

higher-order areas. To better understand the neural mechanisms of auditory scene

analysis, it is essential to understand how the cortical representations of a complex

auditory scene change from the core to the higher order auditory areas.

One topic of interest is whether the brain maintains distinct neural

representations for each unattended source (in addition to the representation of the

attended source), or if all unattended sources are represented collectively as a

single monolithic background object. A common paradigm used to investigate the

neural mechanisms underlying auditory scene analysis employs a pair of speech

streams, of which one is attended, which then leaves the other speech stream

remaining as the background (Kerlin et al., 2010; Ding and Simon, 2012b;

Mesgarani and Chang, 2012; Power et al., 2012; Zion Golumbic et al., 2013b;

O'Sullivan et al., 2015). This results in a limitation, which cannot address the

question of distinct vs. collective neural representations for unattended sources.

This touches on the long-standing debate of whether auditory object segregation is

pre-attentive or it is actively influenced by attention (Carlyon, 2004; Sussman et

al., 2005; Shinn-Cunningham, 2008; Shamma et al., 2011). Evidence for

segregated neural representations of background streams would support the

former, whereas a lack of segregated background objects would support the latter.

To address these issues, we use MEG to investigate a variety of potential

cortical representations of the elements of a multi-talker auditory scene. We test

two major hypotheses: that the dominant representation in core auditory cortex is

of the physical acoustics, not of separated auditory objects; and that once object-

based representations emerge in higher order auditory areas, the unattended

contributions to the auditory scene are represented collectively as a single

background object. The methodological approach employs the linear systems

methods of stimulus prediction and MEG response reconstruction (Lalor et al.,

2009; Mesgarani et al., 2009; Ding and Simon, 2012b; Mesgarani and Chang,

2012; Pasley et al., 2012; Di Liberto et al., 2015).

Materials & Methods:

Subjects & Experimental Design Nine normal-hearing, young adults (6 Female) participated in the experiment. All subjects were paid for their participation. The experimental procedures were approved by the University of Maryland Institutional Review Board. Subjects listened to a mixture of three speech segments spoken by, respectively, a male adult, female adult and a child speaker. The three speech segments were mixed into a single audio channel with equal perceptual loudness. All three speech segments were taken from public domain narration of Grimms' Fairy Tales by Jacob & Wilhelm Grimm (https://librivox.org/fairy-tales-by-the-brothers-grimm/). Periods of silence longer than 300 ms were replaced by a shorter gap whose duration was chosen randomly between 200 ms and 300 ms. The audio signal was low-pass filtered below 4 kHz. In first of three conditions, the subjects were asked to attend to the child speaker, while ignoring the other two (i.e., child speaker as target, with male and female adult speakers as background). In condition two, during which the same mixture was played as in condition one, the subjects were instead asked to attend to the male adult speaker (with female adult and child speakers as background). Similarly, in condition three, the target was switched to the female adult speaker. Each condition was repeated three times successively, producing three trials per condition. The presentation order of the three conditions was counterbalanced

across subjects. Each trial was of 220 s duration, divided into two 110 s sections,

to reduce listener fatigue. To help participants attend to the correct speaker, the

first 30 s of each section was replaced by the clean recording of the target speaker

alone, followed by a 5 s upward linear ramp of the background speakers.

Recordings of this first 35 s of each segment were not included in any analysis. To

further encourage the subjects to attend to the correct speaker, a target-word was

set before each trial and the subjects were asked to count the number of

occurrences of the target-word in the speech of the attended speaker. Additionally,

after each condition, the subject was asked to recount a short summary of the

attended narrative. The subjects were required to close their eyes while listening.

Before the main experiment, 100 repetitions of a 500-Hz tone pip were presented

to each subject to elicit the M100 response, a reliable auditory response occurring

~100 ms after the onset of a tone pip. This data was used check whether any

potential subjects gave abnormal auditory responses, but no subjects were

excluded based on this criterion.

Data recording and pre-processing MEG recordings were conducted using a 160-

channel whole-head system (Kanazawa Institute of Technology, Kanazawa,

Japan). Its detection coils are arranged in a uniform array on a helmet-shaped

surface of the bottom of the dewar, with ~25 mm between the centers of two

adjacent 15.5-mm-diameter coils. Sensors are configured as first-order axial

gradiometers with a baseline of 50 mm; their field sensitivities are 5 fT/$\sqrt{}$ Hz or

better in the white noise region. Subjects lay horizontally in a dimly lit

magnetically shielded room (Yokogawa Electric Corporation). Responses were

recorded with a sampling rate of 1 kHz with an online 200-Hz low-pass filter and

60 Hz notch filter. Three reference magnetic sensors and three vibrational sensors

were used to measure the environmental magnetic field and vibrations. The

reference sensor recordings were utilized to reduce environmental noise from the

MEG recordings using the Time-Shift PCA method (de Cheveigne and Simon,

2007). Additionally, MEG recordings were decomposed into virtual sensors/

components using denoising source separation (DSS) (Särelä and Valpola, 2005;

de Cheveigne and Simon, 2008; de Cheveigne and Parra, 2014), a blind source

separation method that enhances neural activity consistent over trials. Specifically,

DSS decomposes the multichannel MEG recording into temporally uncorrelated

components, where each component is determined by maximizing its trial-to-trial

reliability, measured by the correlation between the responses to the same stimulus

in different trials. To reduce the computational complexity, for all further analysis

the 157 MEG sensors were reduced, using DSS, to 4 components in each

hemisphere. Also, both stimulus envelope and MEG responses were band pass

filtered between 1 – 8 Hz (delta and theta bands), which correspond to the slow

temporal modulations in speech (Ding and Simon, 2012a, b).

Terminology and Notation As specified in the stimulus description, in each

condition the subject attends to one among the three speech streams. The envelope

of attended speech stream is referred to as the 'foreground' and the envelope of

each of the two unattended speech streams is referred to as the 'individual

background'. In contrast, the envelope of the entire unattended part of the

stimulus, comprising *both* unattended speech streams, is referred to as the

'combined background'. The envelope of entire acoustic stimulus or auditory

scene, comprising of all the three speech streams is referred to as the 'acoustic

scene'. Thus, if S_a, S_b, S_c are three speech stimuli, $Env(S_a + S_b + S_c)$ is the

acoustic scene. In contrast, the sum of envelopes of three speech streams,

$Env(S_a) + Env(S_b) + Env(S_c)$, is referred to as the 'sum of streams', and the

two are not mathematically equal: even though both are functions of the same

stimuli, they differ due to the non-linear nature of a signal envelope (the linear

correlation between the acoustic scene and the sum of streams is typically ~0.75).

Neural responses with latencies less than ~85 ms (typically originating

from core auditory areas) are referred to here as 'early neural responses' and

responses with latencies more than ~85 ms (typically from higher-order auditory

areas) (Ahveninen et al., 2011; Okamoto et al., 2011; Steinschneider et al., 2011)

are referred to as 'late neural responses'.

Temporal Response Function In an auditory scene with a single talker, the

relation between MEG neural response and the presented speech stimuli can be

modeled using a linear temporal response function (TRF) as

$$r(t) = \sum_{\tau} s(t - \tau) TRF(\tau) + \varepsilon(t) \qquad (1)$$

where $t = 0,1, \ldots, T$ is time, $r(t)$ is the response from any individual sensor or

DSS component, $s(t)$ is the stimulus envelope in decibels, $TRF(t)$ is the TRF

itself, and $\epsilon(t)$ is residual response waveform not explained by the TRF model

(Ding and Simon, 2012a). The envelope is extracted by averaging the auditory

spectrogram, (Chi et al., 2005) along the spectral dimension. The TRF is estimated

using boosting with 10-fold cross-validation (David et al., 2007). In case of single

speech stimuli, the TRF is typically characterized by a positive peak between 30

ms and 80 ms and a negative peak between 90 ms and 130 ms, referred to as

M50$_{\text{TRF}}$ and M100$_{\text{TRF}}$ respectively (Ding and Simon, 2012b) (positivity/negativity

of the magnetic field is by convention defined to agree with the corresponding

electroencephalography[EEG] peaks). Success/accuracy of the linear model is

evaluated by how well it predicts neural responses, as measured by the proportion

of the variance explained: the square of the Pearson correlation coefficient

between the MEG measurement and the TRF model prediction.

In the case of more than one speaker, the MEG neural response, $r(t)$ can be

modeled as the sum of the responses to the individual acoustic sources (Ding and

Simon, 2012b; Zion Golumbic et al., 2013b), referred to here as the 'Summation

model'. For example, with two speech streams, the neural response would be

modeled as

$$r(t) = \sum_{\tau} S_a(t - \tau) TRF_a(\tau) + \sum_{\tau} S_b(t - \tau) TRF_b(\tau) + \varepsilon(t) \qquad (2)$$

where $S_a(t)$ and $S_b(t)$ are the envelopes of the two speech streams, and $TRF_a(t)$,

and $TRF_b(t)$ are the TRFs corresponding to each stream. The summation model is

easily extended to the case of more than two speech streams, by adding new terms

with each new individual speech stream envelope and the corresponding TRF.

In addition to the existing summation model, we propose a new encoding-

model referred to as the 'Early-late model', which allows one to incorporate the

hypothesis that the early neural responses typically represent the entire acoustic

scene, but that the later neural responses differentially represent the separated

foreground and background.

$$r(t) = \sum_{\tau=0}^{\tau=\tau_1} S_A(t - \tau) TRF_A(\tau) + \sum_{\tau=\tau_1}^{\tau=\tau_2} S_F(t - \tau) TRF_F(\tau) + \sum_{\tau=\tau_1}^{\tau=\tau_2} S_B(t - \tau) TRF_B(\tau) + \epsilon(t) \qquad (3)$$

where $S_A(t)$ is the (entire) acoustic scene, $S_F(t)$ is the envelope of attended

(foreground) speech stream, and $S_B(t)$ is the combined background (i.e., envelope

of everything other than attended speech stream in the auditory scene), and

$TRF_A(t), TRF_F(t),$ and $TRF_B(t)$ are the corresponding $TRFs$. τ_1, τ_2 represent the

boundary values of the integration windows for early and late neural responses

respectively.

The explanatory power of different models, such as the Summation and

Early-late models, can be ranked by comparing the accuracy of their response

predictions (illustrated in Figure 1, left).

240   (Figure 1 about here)

Decoding speech from neural responses While the TRF/encoding analysis

described in the previous section predicts neural response from the stimulus,

decoding analysis reconstructs the stimulus based on the neural response. Thus,

decoding analysis complements the TRF analysis (Mesgarani et al., 2009).

Mathematically the envelope reconstruction/decoding operation can be formulated

as

$$E(t) = \sum_{k=1}^{N} \sum_{\tau=\tau_b}^{\tau_e} M_k(t+\tau) D_k(\tau) + \epsilon(t) \qquad (4)$$

where $E(t)$ is the reconstructed envelope, $M_k(t)$ is the MEG recording (neural

response) from sensor/component k, and $D_k(t)$ is the linear decoder for

sensor/component k. The times τ_b and τ_e denote the beginning and end times of

the integration window. By appropriately choosing the values of τ_b and τ_e,

envelope reconstructions using neural responses from any desired time window

can be compared. The decoder is estimated using boosting analogously to the TRF

estimation in the previous section. In the single talker case the envelope is of that

talker's speech. In a multi-talker case, the envelope to be reconstructed might be

the envelope of the speech of attended talker, or one of the background talkers, or

of a mixture of any two or all three talkers, depending on the model under

consideration. Chance-level reconstruction (i.e., the noise floor) from a particular

neural response is estimated by reconstructing an unrelated stimulus envelope

from that neural response. Figure 2 illustrates the distinction between

reconstruction of stimulus envelope from early and late responses. The stimulus

envelope at time point t can be reconstructed using neural responses from the

dashed (early response) window or dotted (late response) window. (While it is true

that the late responses to the stimulus at time point $t - \Delta t$ overlap with early

responses to the stimulus at time point t, the decoder used to reconstruct the

stimulus at time point t from early responses is only minimally affected by late

responses to the stimulus at time point $t - \Delta t$ when the decoder is estimated by

averaging over a long enough duration, e.g., tens of seconds). The cut-off time

between early and late responses, $\tau_{boundary}$, was chosen to minimize the overlap

between the M50$_{TRF}$ and M100$_{TRF}$ peaks, on a per subject basis, with a typical

value being 85 ms. When decoding from early responses only, the time window of

integration is from $\tau_b = 0$ to $\tau_e = \tau_{boundary}$. When decoding from late neural

responses only, the time window of integration is from $\tau_b = \tau_{boundary}$ to $\tau_e =$

500 ms.

277   (Figure 2 about here)

The robustness of different representations, such as of Foreground vs.

Background, can be compared by examining the accuracy of their respective

stimulus envelope reconstructions (illustrated in Figure 1, right).

Statistics All statistical comparisons reported here are two-tailed permutation tests

with N=1,000,000 random permutations (within subject). Due to the value of N

selected, the smallest accurate p value that can be reported is $2\times1/N$ ($= 2\times10^{-6}$; the

factor of 2 arises from the two-tailed test) and any p value smaller than $2/N$ is

reported as $p < 2\times10^{-6}$. The statistical comparison between foreground and

individual backgrounds requires special mention, since each listening condition

has one foreground but two individual backgrounds. From the perspective of both

behavior and task, both the individual backgrounds are interchangeable. Hence,

when comparing reconstruction accuracy of foreground vs. individual background

the average reconstruction accuracy of the two individual backgrounds is used.

Finally, Bayes factor analysis is used, when appropriate, to evaluate evidence in

favor of null hypothesis, since conventional hypothesis testing is not suitable for

such purposes. Briefly, Bayes factor analysis calculates the *posterior odds* i.e., the

ratio of $P(H_0|observations)$ to $P(H_1|observations)$, where H_0 and H_1 are the null

and alternate hypotheses respectively.

$$\frac{P(H_0|observations)}{P(H_1|observations)} = \frac{P(observations|H_0)}{P(observations|H_1)} \times \frac{P(H_0)}{P(H_1)} \qquad (5)$$

$$= BF_{01} \times \frac{P(H_0)}{P(H_1)} \qquad (6)$$

The ratio of $P(observations|H_0)$ and $P(observations|H_1)$ is denoted as the Bayes

factor, BF_{01}. Then, under the assumption of equal priors ($P(H_0) = P(H_1)$), the

posterior odds reduces to BF_{01}. A BF_{01} value of 10 indicates that the data is ten

times more likely to occur under the null hypothesis than the alternate hypothesis;

conversely, a BF_{01} value of 0.1 indicates that the data is 10 times more likely to

occur under the alternate hypothesis than the null hypothesis. Conventionally, a

BF_{01} value between 3 and 10 is considered as moderate evidence in favor of the

null hypothesis, and a value between 10 and 30 is considered strong evidence;

conversely, a BF_{01} value between 1/3 & 1/10 (respectively 1/10 & 1/30) is

considered moderate (respectively strong) evidence for the alternate hypothesis

(for more details we refer the reader to Rouder et al. (2009)).

Results

Stimulus reconstruction from early neural responses

To investigate the neural representations of the attended vs. unattended speech

streams associated with early auditory areas, i.e., from core auditory cortex,

(Nourski et al., 2014), the temporal envelope of attended (foreground) and

unattended speech streams (individual backgrounds) were reconstructed using

decoders optimized individually for each speech stream. All reconstructions

performed significantly better than chance level (foreground vs. noise, $p < 2\times10^{-6}$;

individual background vs. noise, $p < 2\times10^{-6}$), indicating that all three speech

streams are represented in early auditory cortex. Figure 3A shows reconstruction

accuracy for foreground vs. individual backgrounds. A permutation test shows no

significant difference between foreground and individual background ($p = 0.21$),

indicating that there is no evidence of significant neural bias for the attended

speech stream over the ignored speech stream, in early neural responses. In fact,

Bayes Factor analysis ($BF_{01} = 4.2$) indicates moderate support in favor of the null

hypothesis (Rouder et al., 2009), that early neural responses do not distinguish

significantly between attended and ignored speech streams.

328   (Figure 3 about here)

To test the hypothesis that early auditory areas represent the auditory scene

in terms of acoustics, rather than as individual auditory objects, we reconstructed

the acoustic scene (the envelope of the sum of all three speech streams) and

compared it against the reconstruction of the sum of streams (sum of

reconstruction envelopes of each of the three individual speech streams). Separate

decoders optimized individually were used to reconstruct the acoustic scene and

the sum of streams. As can be seen in Figure 3B, the result shows that the acoustic

scene is better reconstructed than the sum of streams ($p < 2\times10^{-6}$). This indicates

that early auditory cortex is better described as processing the entire acoustic scene

rather than processing the separate elements of the scene individually.

Stimulus reconstruction from late neural responses

While the preceding results were based on early cortical processing, the following

results are based on late auditory cortical processing (responses with latencies

more than ~85 ms). Figure 4A shows the scatter plot of reconstruction accuracy

345 for the foreground vs. individual background envelopes based on late responses. A

346 paired permutation test shows that reconstruction accuracy for the foreground is

347 significantly higher than the background ($p < 2 \times 10^{-6}$). Even though the individual

348 backgrounds are not as reliably reconstructed as foreground, their reconstructions

349 are nonetheless significantly better than chance level ($p < 2 \times 10^{-6}$).

350   In order to distinguish among possible neural representations of the

351 background streams, we compared the reconstructability of the envelope of the

352 entire background as a whole, with the reconstructability of the sum of the

353 envelopes of the (two) backgrounds. If the background is represented as a single

354 auditory object (i.e., "the background"), the reconstruction of the envelope of the

355 entire background should be more faithful than the sum of envelopes of individual

356 backgrounds. In contrast, if the background is represented as distinct auditory

357 objects, each distinguished by its own envelope, the reconstruction of the sum of

358 envelopes of the individual backgrounds should be more faithful. Figure 4B shows

359 the scatter plot of reconstruction accuracy for the envelope of combined

360 background vs. the sum of the envelopes of the individual background streams.

361 Analysis shows that the envelope of the combined background is significantly

362 better represented than the sum of the individual envelopes of the individual

363 backgrounds ($p = 0.012$). As noted previously, the envelope of the combined

364 background is actually strongly correlated with the sum of the envelopes of the

individual backgrounds, meaning that finding a significant difference in their

reconstruction accuracy is *a priori* unlikely, providing even more credence to the

result.

(Figure 4 about here)

Encoding analysis

Results above from envelope reconstruction suggest that while early neural

responses represent the auditory scene in terms of the acoustics, the later neural

responses represent the auditory scene in terms of a separated foreground and a

single background stream. In order to further test this hypothesis, we use TRF-

based encoding analysis to directly compare two different models of auditory

scene representations. The two models compared are the standard Summation

model (based on parallel representations of all speech streams; see Equation 2) and

the new Early-late model (based on an early representation of the entire acoustic

scene and late representations of separated foreground and background; see

Equation 3). Figure 5 shows the response prediction accuracies for the two

models. A permutation test shows that the accuracy of the Early-late model is

considerably higher than that of the Summation model ($p < 2 \times 10^{-6}$). This indicates

that a model in which early/core auditory cortex processes the entire acoustic

scene but later/higher-order auditory cortex processes the foreground and

background separately has more support than the previously employed model of

parallel processing of separate streams throughout auditory cortex.

389   (Figure 5 about here)

Discussion

In this study, we used cortical tracking of continuous speech, in a multi-talker

scenario, to investigate the neural representations of an auditory scene. Differing

latencies of the neural sources processing the same stimuli allow us to separate the

source activity temporally, thus enabling the tracking of differing neural

representations of the auditory scene. From MEG recordings of subjects

selectively attending to one of the three co-located speech streams, we observed

that 1) The early neural responses (with short latencies), which originate primarily

from core auditory cortex, represent the foreground (attended) and background

(ignored) speech streams without any significant difference, whereas the late

neural responses (with longer latencies), which originate primarily from higher-

order areas of auditory cortex, represent the foreground with significantly higher

fidelity than the background; 2) Early neural responses are not only balanced in

how they represent the constituent speech streams, but in fact represent the entire

acoustic scene holistically, rather than as separately contributing individual

perceptual objects; 3) Even though there are two physical speech streams in the

background, no neural segregation is observed for the background speech streams.

It is well established that auditory processing in cortex is performed in a

hierarchical fashion, in which an auditory stimulus is processed by different

anatomical areas at different latencies (Inui et al., 2006; Nourski et al., 2014).

Using this idea to inform the neural decoding/encoding analysis allows the

effective isolation of neural signals from a particular cortical area, and thereby the

ability to track changes in neural representations as the stimulus processing

proceeds along the auditory hierarchy. This time-constrained

reconstruction/prediction approach may prove especially fruitful in high-time-

resolution/low-spatial-resolution imaging techniques such as MEG and EEG. Even

though different response components are generated by different neural sources,

standard neural source localization algorithms may perform poorly when different

sources are strongly correlated in their responses (Lutkenhoner and Mosher,

2007). While the proposed method is not to be viewed as an alternative to source

localization methods, it can nonetheless be used to tease apart different

components of MEG/EEG response, without explicit source localization.

The envelope reconstruction using the early, auditory core, neural response

component showed no significant difference between foreground and background,

in contrast to reconstruction using the late, higher-order auditory, neural

responses, where the foreground is substantially better represented than any

individual background. This *decoding* result is in agreement with the *encoding*

result of (Ding and Simon, 2012b) where the authors showed that the early

$M50_{TRF}$ component of the temporal response function is not significantly

modulated by attention, whereas the late $M100_{TRF}$ component is modulated by

attention.

Even though there is no significant difference between the ability to

reconstruct the foreground and background from early neural responses,

nonetheless we observe a non-significant tendency towards an enhanced

representation of the foreground (foreground > background, $p = 0.21$). This could

be due to task-related plasticity of spectro-temporal receptive fields of neurons in

mammalian primary auditory cortex (Fritz et al., 2003), where the receptive fields

of neurons are tuned to match the stimulus characteristics of attended sounds. It

could also be explained by entrainment (Schroeder and Lakatos, 2009; Zion

Golumbic et al., 2012), which postulates that the high excitability periods of

neurons become aligned with temporal structure of foreground, thereby enhancing

its neural representation.

The increase in fidelity of the foreground as the response latency increases,

from early neural responses (from core auditory cortex) to late neural responses

(from higher-order auditory cortex), indicates a temporal as well as functional

hierarchy in cortical processing of auditory scene, from core to higher-order areas

in auditory cortex. Similar preferential representation for the attended speech

stream has been demonstrated, albeit with only two speech streams, using delta

and theta band neural responses (Ding and Simon, 2012b; Zion Golumbic et al.,

2013a; Zion Golumbic et al., 2013b) as well as high-gamma neural responses

(Mesgarani and Chang, 2012; Zion Golumbic et al., 2013a), and using monaural

(Ding and Simon, 2012b; Mesgarani and Chang, 2012) as well as audio-visual

speech (Zion Golumbic et al., 2013a; Zion Golumbic et al., 2013b).

While some researchers suggest a selective entrainment model (Schroeder

and Lakatos, 2009; Ng et al., 2012; Zion Golumbic et al., 2013b; Kayser et al.,

2015) as the mechanism underlying the selective tracking of attended speech,

others suggest a temporal coherence model (Shamma et al., 2011; Ding and

Simon, 2012b) as the neuronal mechanism underlying selective tracking. Natural

speech is quasi-rhythmic with different dominant rates at syllabic, word and

prosodic frequencies. The selective entrainment model suggests that attention

causes endogenous low frequency neural oscillations to align with the temporal

structure of the attended speech stream, thus aligning the high excitability phases

of oscillations with events in attended stream. This effectively forms a mask that

favors the attended speech. The temporal coherence model suggests that selective

tracking of attended speech is achieved through two stages. First is a cortical

filtering stage, where feature selective neurons filter the stimulus producing a

multidimensional representation of auditory scene along different feature axes.

This is followed by a second stage, coherence analysis, which combines different

features streams based on their temporal similarity, giving rise to separate

perceptions of attended and ignored streams.

The representation of an auditory scene in core auditory cortex is here

shown to be more spectro-temporal- or acoustic-based than object-based, as

demonstrated by the result that the envelope of the auditory scene is better

reconstructed than the sum of envelopes of the individual speech streams (e.g.,

Figure 3B). This is further supported by the result that the Early-late model

predicts MEG neural responses significantly better than Summation model (e.g.,

Figure 5). This is consistent with previous studies that demonstrated that neural

activity in core auditory cortex was highly sensitive to acoustic characteristics of

speech and primarily reflects spectro-temporal attributes of sound (Nourski et al.,

2009; Okada et al., 2010; Steinschneider et al., 2014). All these results suggest that

early neural responses, primarily from core auditory cortex, reflect an acoustic-

based representation rather than object-based. In contrast, Nelken and Bar-Yosef

(2008) suggest that neural auditory objects may form as early as primary auditory

cortex, and Fritz et al. (2003) show that representations of dynamic sounds in

primary auditory cortex are influence by task. It is possible that less complex

stimuli are resolved earlier in the hierarchy of auditory pathway (e.g., sounds that

can be separated via tonotopy) whereas speech streams, which overlap both

spectrally and temporally, are resolved only much later in auditory pathway.

It is widely accepted that an auditory scene is *perceived* in terms of

auditory objects (Bregman, 1994; Griffiths and Warren, 2004; Shinn-Cunningham,

2008; Shamma et al., 2011). Ding and Simon (2012a) demonstrated evidence for

an object-based cortical representation of an auditory scene, but did not distinguish

between early and late neural responses. This, coupled with the result here that

early neural responses provide an acoustic, not object-based, representation,

strongly suggest that the object-based representation emerges only in the late

neural responses/higher-order (belt and parabelt) auditory areas. This is further

supported by the observation that acoustic invariance, a property of object-based

representation, is observed in higher order areas but not in core auditory cortex

(Chang et al., 2010; Okada et al., 2010).

When the foreground is represented as an auditory object in late neural

responses, the finding that the combined background is better reconstructed than

the sum of envelopes of individual backgrounds (Figure 4B) suggests that in late

neural responses the background is not represented as separated and distinct

auditory objects. This result is consistent with that of Sussman et al. (2005), who

reported an unsegregated background when subjects attended to one of three tone

streams in the auditory scene. This unsegregated background may be a result of an

'analysis-by-synthesis' (Yuille and Kersten, 2006; Poeppel et al., 2008)

mechanism, wherein the auditory scene is first decomposed into basic acoustic

elements, followed by top-down processes that guide the synthesis of the relevant

components into a single stream, which then becomes the object of attention. The

remainder of the auditory scene would be the unsegregated background, which

itself might have the properties of an auditory object. When attention shifts, new

auditory objects are correspondingly formed, with the old ones now contributing

to the unstructured background. Shamma et al. (2011) suggest that this top down

influence acts through the principle of temporal coherence. Between the two

opposing views, that streams are formed pre-attentively and that multiple streams

can co-exist simultaneously, or that attention is required to form a stream and only

that single stream is ever present as separated perceptual entity, these findings lend

support to the latter.

In summary, these results provide evidence that, in a complex auditory

scene with multiple overlapping spectral and temporal sources, the core areas of

auditory cortex maintains an acoustic representation of the auditory scene with no

significant preference to attended over ignored source, and with no separation into

distinct sources. It is only the higher-order auditory areas that provide an object

based representation for the foreground, but even there the background remains

unsegregated.

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 Legend:

 Figure 1: Illustrations of different decoding- and encoding-based neural

 representations of the auditory scene and its constituents. (*Left)* Examples of

 predicted MEG neural response using the Early-late model (red) and the

 Summation model (magenta) superimposed on actual MEG response (black). The

 proposed Early-late model prediction shows higher correlation with the actual

 MEG neural response than Summation model. (*Right)* Example of speech

 envelopes reconstructed (grey) from their late neural responses, for both the

 foreground and the background, superimposed on actual speech envelopes of

 foreground (blue) and background (cyan). The foreground reconstruction shows

 higher correlation with the actual foreground envelope, compared to the

 background reconstruction with the actual background envelope. All examples are

 grand averages across subjects (3 seconds duration).

 Figure 2: Early vs. late MEG neural responses to a continuous speech stimulus. A

 sample stimulus envelope and multi-channel MEG recordings are shown in red

 and black respectively. The two grey vertical lines indicate two arbitrary time

 points at $t - \Delta t$ and t. The dashed and dotted boxes represent the early and late

 MEG neural responses to stimulus at time point t respectively. The reconstruction

of the stimulus envelope at time t can be based on either early or late neural

responses, and the separate reconstructions can be compared against each other.

Figure 3: Stimulus envelope reconstruction accuracy using *early* neural responses.

A. Scatter plot of reconstruction accuracy of the foreground vs. individual

background envelopes. No significant difference was observed ($p = 0.21$), and

therefore no preferential representation of the foreground speech over the

individual background streams is revealed in early neural responses. **B.** Scatter

plot of reconstruction accuracy of the envelope of the entire acoustic scene vs. that

of the sum of the envelopes of all three individual speech streams. The acoustic

scene is reconstructed more accurately (visually, most of data points fall above the

diagonal) as a whole than as the sum of individual components in early neural

responses ($p < 2 \times 10^{-6}$). Reconstruction accuracy is measured by proportion of the

variance explained: the square of the Pearson correlation coefficient between the

actual and predicted envelopes.

Figure 4: Stimulus envelope reconstruction accuracy using *late* neural responses.

A. Scatter plot of accuracy between foreground vs. individual background

envelope reconstructions demonstrates that the foreground is represented with

dramatically better fidelity (visually, most of data points fall above the diagonal)

than the background speech, in late neural responses ($p < 2 \times 10^{-6}$). **B.** Scatter plot

of the reconstruction accuracy of the envelope of the entire background vs. that of

the sum of the envelopes of the two individual background speech streams. The

background scene is reconstructed more accurately as a monolithic background

than as separated individual background streams in late neural responses ($p =$

0.012)

Figure 5: MEG response prediction accuracy. Scatter plot of the accuracy of

predicted MEG neural response for the proposed Early-late model vs. the standard

Summation model. The Early-late model predicts the MEG neural response

dramatically better (visually, most of data points fall above the diagonal) than the

Summation model ($p < 2 \times 10^{-6}$). The accuracy of predicted MEG neural

responses is measured by proportion of the variance explained: the square of the

Pearson correlation coefficient between the actual and predicted responses.

Neural response prediction from stimulus **Stimulus reconstruction from neural response**

— Neural response
- - - Summation model
- - - Early-late model

— Foreground stimulus envelope
— Background stimulus envelope
- - - Reconstructed envelope

718   Figure 1

721   Figure 2

Stimulus Reconstruction Accuracy from **Early** Neural Responses

724   Figure 3

Stimulus Reconstruction Accuracy from **Late** Neural Responses

727   Figure 4

730   Figure 5