Formants from the Wave Equation and Recording Speech During MRI

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Goals

- The main goal is to simulate vowels based on a wave equation model.
- In the relatively near future, this simulator could be used to plan oral and maxillofacial surgery and to investigate abnormal anatomy.
- In the long term, we would like to predict the effects that medical operations will have on a patient's speech production.

Background



Figure 1: (a) The wave equation model and (b) a sample vowel geometry

• We need an extensive amount of additional accurate anatomic data and simultaneously recorded sound to validate the simulation results.

Mathematics: Finite Element Method in our project



- Our initial mesh had about 64000 tetrahedral elements.
- We do not aim to produce real time synthesis using the wave equation but we do expect to get reasonably close.
- The corresponding resonance problem, i.e., the computation of formants is not so heavy numerically.

Results: Pressure distributions



Figure 2: Four approximate eigenfunctions corresponding to the lowest eigenvalues ie. pressure distributions for formants 1-4. Glottis is on the left and mouth on the right.

Sound measurements: What would we like to get?

- The fundamental frequency F0, ...
- F1, F2, F3 and, if possible, F4 ...
- ... and their bandwidths ...
- ... before, after and during the MR imaging sequence.
- Access to clean speech signal in real time.

We would like to obtain large amounts of high quality, anatomically and phonetically relevant, simultaneous image and sound data.

Sound measurements: What's the problem then?

- No metal allowed inside the MRI main coil.
- No ferromagnetic material allowed inside the MRI room.
- All electronics in the MRI room have to be RF-shielded.
- Strong acoustic noise (over 90 dB SPL) present during the imaging sequence.

Sound collector

We have constructed a two channel sound collector for our system. One channel will sample the noise and the other the contaminated speech.



(a)

(b)

Figure 3: The sound collector from (a) below with the sound source and (b) above

Faraday cage





(b)

Figure 4: (a) The Faraday cage houses the microphones and (b) the acoustic waveguides enter the cage through electromagnetic waveguides

Tests: Does the noise cancellation work with acoustic components?



Figure 5: CMRR of the whole system excluding the sound collector

Thank you. Questions, please?