

Speech Coding Based on Physiological Models of Speech Production

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(from *Advances in Speech Signal Processing*,
Furui & Sondhi, eds.)

Discussion led by Tina Bennett

Outline

- Motivation (*Section 1* - 1 slide)
- Model Components (*Section 2* - 6 slides)
- Deriving Parameters (*Section 3* - 4 slides)
- Results and Future Directions – circa 1992
(*Sections 3.3 & 4* – 1 slide)

Motivation Arguments

- No source-filter model → instead, interaction between *excitation* and *impedence* (vocal/nasal tract)
- Thus, more natural speech produced
→ better parameters for coding, pitch changes natural and automatic, interpolation of parameters

Model components

2.1 Geometry of Vocal & Nasal Tracts

- *Vocal Tract: 25 (now 35) years of articulatory modeling (x-ray data)*
 - Their own work based on Mermelstein (1973)
 - see Figure 1 and Table 1 for parameters
- *Nasal Tract: fixed filter is sufficient*
 - Nasal tract w/model of sinus maxillaries can be efficiently modeled by Helmholtz resonator coupled to nasal tract

Model components

2.2 *Wave Propagation in the Tracts*

- Assumptions:
 - *linear wave equation applies*
 - exception: just before & after plosives
 - *tract can be approximated as variable-area tube*
 - but what effects of curvature?
(very little for tube w/*constant* radius)
 - *the motion is planar*
 - reasonable for 3500 Hz or less (tube opening 1.9 inches max), but otherwise?

Model components

2.2 *Wave Propagation in the Tracts*

- Use *chain matrix* to compute acoustical properties of the tube:
 - relates pressure & volume velocities at one end of the tube to the other (i.e. $P_{\text{out}}, U_{\text{out}}$ - lips/nose end; $P_{\text{in}}, U_{\text{in}}$ - glottis end)
- See matrix formula, p. 237
 - Tube is approximated by concatenating 10-20 constant segments (easier to compute K for constant cross section)
 - Can account for losses and walls (see details in paper)
 - Note that all linear properties can be derived (transfer function, input impedance) → equations 2 & 3, p. 238

Model components

2.3 Modeling the Excitation

- Two types (ignoring clicks)
 - *Voiced excitation*: lung air flow interrupted by glottal vibration (i.e. voiced sounds)
 - *Turbulent excitation*: flow through narrow constriction (i.e. fricatives & aspiration)
 - May be combined (i.e. voiced fricatives)

Model components

2.3 Modeling the Excitation

- Modeling Voiced Excitation
 - Two-mass model (Ishizaka & Flanagan, 1972)
 - Figure 2, p. 239
 - Equations 4 through 8b (pp. 239-240)
 - Improvements from incorporation of additional losses (e.g. vertical phasing)
 - Lung pressure assumed to be constant (no breathy voice!)
 - Glottal “chink” (affects F_0 ; helpful for /h/-to-vowel transitions)
 - Try parametric model of glottal area function instead
 - Advantages: independent control of acoustic features; precise positioning of glottal closures
 - Disadvantages: need more parameters & higher update rate

Model components

2.3 Modeling the Excitation

- Modeling Turbulent Excitation
 - Frication (constriction in vocal tract)
 - Figure 3, equations 10-13, pp. 242-243
 - Series noise location critical (different depending on which fricative), but not for volume velocity source – equation 14, p. 243
 - Aspiration (turbulence at glottis)
 - Same approach

Deriving parameters

- Note 1 to many mapping problem (derivation from LPC vector fails here)
- Their approach: (Figure 4, p. 247)
 - Analysis-by-synthesis & Hooke-and-Jeeves optimization methods
 - Parameters iteratively adjusted to minimize cost function

Deriving parameters

3.1.1 More about the Cost Function

- Cost function: evaluate match between signals (original & synthetic)
- For cost function, compute the following: (for original and synthetic speech)
 - autocorrelations normalized by residual energy
 - tenth-order LPC vectors
 - autocorrelations of LPC vectors
- Alleviates non-uniqueness problem

Deriving parameters

3.1.1 More about the Cost Function

- Four components (3 similarity factors, 1 smoothness penalization): (see equations 17-23, pp. 249-250)
 - 1) symmetrized likelihood ratio between LPC vectors (original and synthesized)
 - 2) comparison of energy (original and synthesized)
 - 3) comparison of time derivatives of glottal excitation (original and synthesized)
 - 4) parameter distance between frames: smoothness constraint (penalizes large changes in movement)
- These are combined and weighted (based on voicing)
- Joint optimization (tract & glottal params.) worked best

Deriving parameters

3.2 Initialization of Tract Optimization

- Problem: will converge to local (possibly bad) minimum
- Solution: use a codebook for starting shape of vocal tract
 - must cover space of natural speech
 - must ignore spectral tilt
 - see Figure 5 for sample comparison of formant spaces (natural speech vs. using codebook)

Results & Future (*circa 1992*)

- Figure 7: original utterance compared with two synthesized versions
 - optimizing parameters of Mermelstein articulatory model & optimizing tract areas
 - area optimization wins
- Benefits of articulatory model (e.g. interpolation) argue for further investigation
 - improve model for acoustic properties from tracts (incorporate variable losses, more than one fricative noise source)
 - parametric models for glottal opening
 - strategies for coding glottal and tract parameters

Note: References are not noted
here; please see the paper!