Measuring the Perceptual Effects of Speech Synthesis Modelling Assumptions

Rech Speech Technology

Edinburgh - Cambridge - Sheffield

Gustav Eje Henter, Thomas Merritt, Matt Shannon, Catherine Mayo, Simon King



Summary

"Hear the perceptual effects of modelling assumptions in statistical speech synthesis"



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"Hear the perceptual effects of modelling assumptions in statistical speech synthesis"

1. Through manipulating repeated natural speech



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"Hear the perceptual effects of modelling assumptions in statistical speech synthesis"

- 1. Through manipulating repeated natural speech
- 2. Identify which assumptions that limit synthesiser naturalness



Overview

- 1. Background
- 2. Methodology
- 3. Experiments
- 4. Conclusions and outlook



Naturalness in speech synthesis

Output naturalness depends on many factors:

- Text processing
- Speech parameter representation (vocoder etc.)
- Probabilistic models
- Parameter generation method



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- High-level assumptions
 - Different parameter streams are conditionally independent
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Assumption adequacy affects output naturalness



Questions

- 1. Which high-level assumptions hurt naturalness?
- 2. How much may we gain if we could remove these assumptions?



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- 2. How much may we gain if we could remove these assumptions?
- $\rightarrow\,$ Where should we direct our improvement efforts?



Traditional fault-finding

Investigate naturalness through trial-and-error:

- 1. Select an assumption and modify it
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 - Low-level assumptions
 - Estimation errors



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 - Estimation errors
- Does not compare the relative severity of different assumptions



Our insight

• Natural speech is a sample from the true acoustic model



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- By manipulating repeated natural speech we can simulate output from
 - highly accurate models
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- Natural speech is a sample from the true acoustic model
- By manipulating repeated natural speech we can simulate output from
 - highly accurate models
 - only incorporating certain high-level modelling assumptions
 - no low-level assumptions at all
 - $\circ~$ with a particular parameter representation
 - $\circ~$ and a particular output generation method



Why is this cool?

Nobody knows what these "nearly perfect" models are, yet we can listen to their output!



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Nobody knows what these "nearly perfect" models are, yet we can listen to their output!

- Compare naturalness degradations due to different high-level assumptions in an otherwise perfect model
- Identified key naturalness bottlenecks in speech synthesis



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REHASP 0.5 corpus

- "REpeated HArvard Sentence Prompts"
- Female British English talker "Lucy"
- 30 Harvard sentence prompts
- Each read aloud 40 times
 - Presented in random order
- Recorded at 16 bit 96 kHz
- Publicly available under a permissive license
 - datashare.is.ed.ac.uk/handle/10283/561



0. Start with natural speech repetitions:





1. Extract parameters:









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Speech representation

Standard parametric speech representation used for experiments:

- 16 kHz operating point
- Matlab STRAIGHT for parameter extraction
- 46-dimensional parameter vector with three streams:
 - 40 MCEPs (0-39), representing filter coefficients
 - Log-F0
 - 5 band aperiodicities (BAPs)
- 5 ms frame shift



1.b. Resynthesise (baseline "V"):











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In pictures

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2.a. Match frames:









2.a. Match frames:









2.a. Match frames:







2.a. Match frames:





2.b. Warp timings:





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2.c. Resynthesise (baseline "D"):









2.d. Remove reference:









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We now have "LEGO pieces" of aligned repetitions







3.a. Combine parameters from independent repetitions:







3.a. Combine parameters from independent repetitions:







3.a. Combine parameters from independent repetitions:







3.a. Combine parameters from independent repetitions:







3.a. Combine parameters from independent repetitions:







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3.a. Combine parameters from independent repetitions:





3.a. Resynthesise chimeric speech (here condition "SF"):





3.b. Take the mean of all repetitions:







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3.b. Take the mean of all repetitions:





3.b. Resynthesise mean speech (condition "M"):





Interpretation

- Repeated speech \approx independent samples from a "perfect" acoustic model
- Chimeric speech \approx samples from a model making certain high-level assumptions but no low-level assumptions
- Mean speech \approx the mean of a probabilistic model



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Present investigation

- Two model assumption classes:
 - 1. Stream independence assumptions
 - $1.1\,$ Source and filter parameters independent
 - $1.2\;$ Filter, pitch, aperiodicities independent
 - 2. Independence assumptions among filter coefficients



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- = 12 conditions (4 baselines)
 - $\circ~$ For each of the 30 Harvard sentences



What it sounds like

Sampling-based generation:

Database examples:	3	7	26	32	
Baselines:	Ν	VU	V	D	
Stream independence:	SF	SI			
Filter coefficient independence:	L1	L2	H1	H2	

 $\begin{array}{c} \mbox{Mean-based generation:} \\ \mbox{Averaging:} & \mbox{M} \end{array}$

(Also available online at homepages.inf.ed.ac.uk/ghenter)

Reference to the second second

Naturalness test

MUSHRA test for parallel, fine-grained naturalness assessment



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Naturalness results

Box plot of 549 comparisons rating natural speech at 100:



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 - 1. Source-filter independence assumption reduces naturalness
 - 2. Independence assumptions among filter coefficients further reduces naturalness
- Using mean-based parameter generation:
 - 1. Better than sampling for poor models
 - 2. Less natural than sampling for accurate models



Limitations

Conclusions not applicable to:

- Other speech representations
- Other parameter generation methods
 - $\circ~$ E.g., postfiltering, global variance modelling



Future work

• Record REHASP 1.0 corpus



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- Expanded investigation
 - Consider additional assumptions
 - Cover the entire spectrum from natural speech to TTS system
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- Record REHASP 1.0 corpus
- Expanded investigation
 - Consider additional assumptions
 - Cover the entire spectrum from natural speech to TTS system
 - Consider additional parameter generation methods
- Effect of different parameter representations

The end



Thank you for listening!