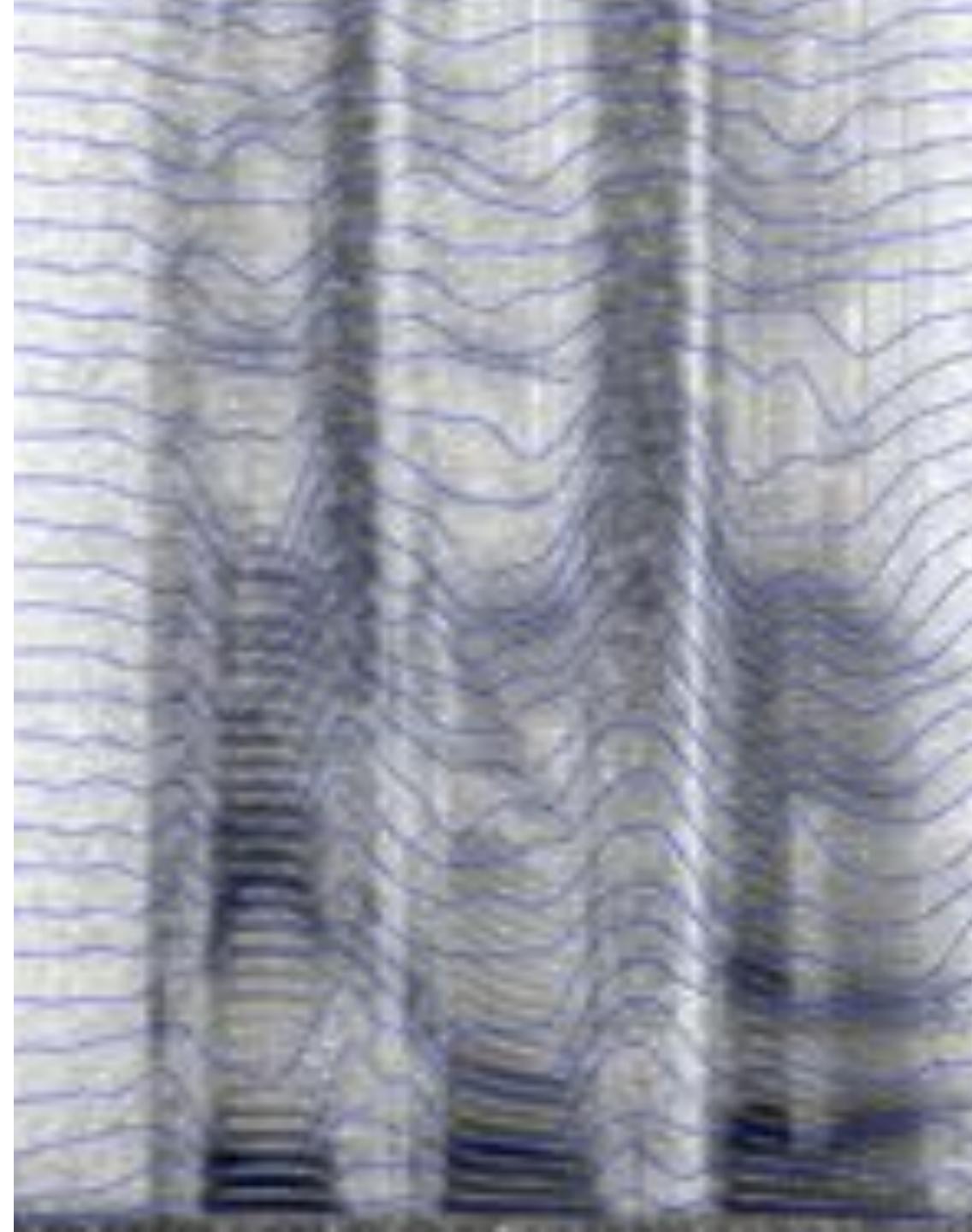
Speech Synthesis

Simon King University of Edinburgh

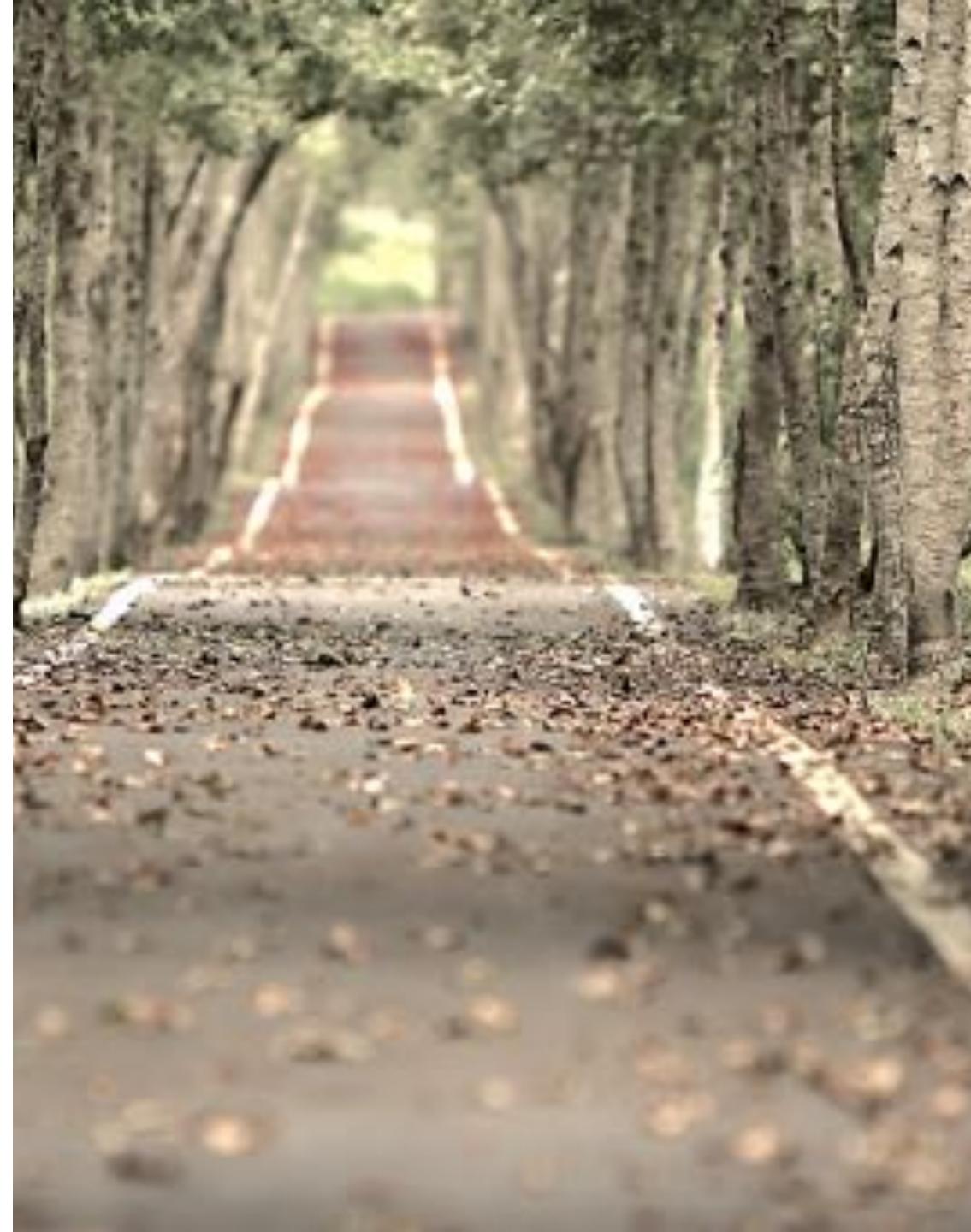


Databases for speech synthesis

- key concepts
- script design
- annotating the database

What you should already know

- the front end
 - linguistic specification
- unit selection method
 - select units from similar contexts
 - target cost measures this similarity
- basic Automatic Speech Recognition
 - Hidden Markov Models
 - finite state language model
 - decoding





Databases for speech synthesis

- <u>key concepts</u>
- script design
- annotating the database

Key concept: base unit type

- relatively small number of types
 - e.g., diphone
- in unit selection
 - base unit type is strictly matched between target and candidate
- therefore, target cost does **not** need to query the base unit type
 - only query its context

• unless database is badly designed: then we would have to back off to a similar type

Key concept: context

- the linguistic and acoustic environment in which a base unit occurs, including • **phonetic context** - the sounds before and after it
- - prosodic environment stress, prosody, ...
 - **position** in the syllable, word, phrase, ...
- Exact features considered will depend on target cost formulation (e.g., IFF or ASF)

Key concept: coverage

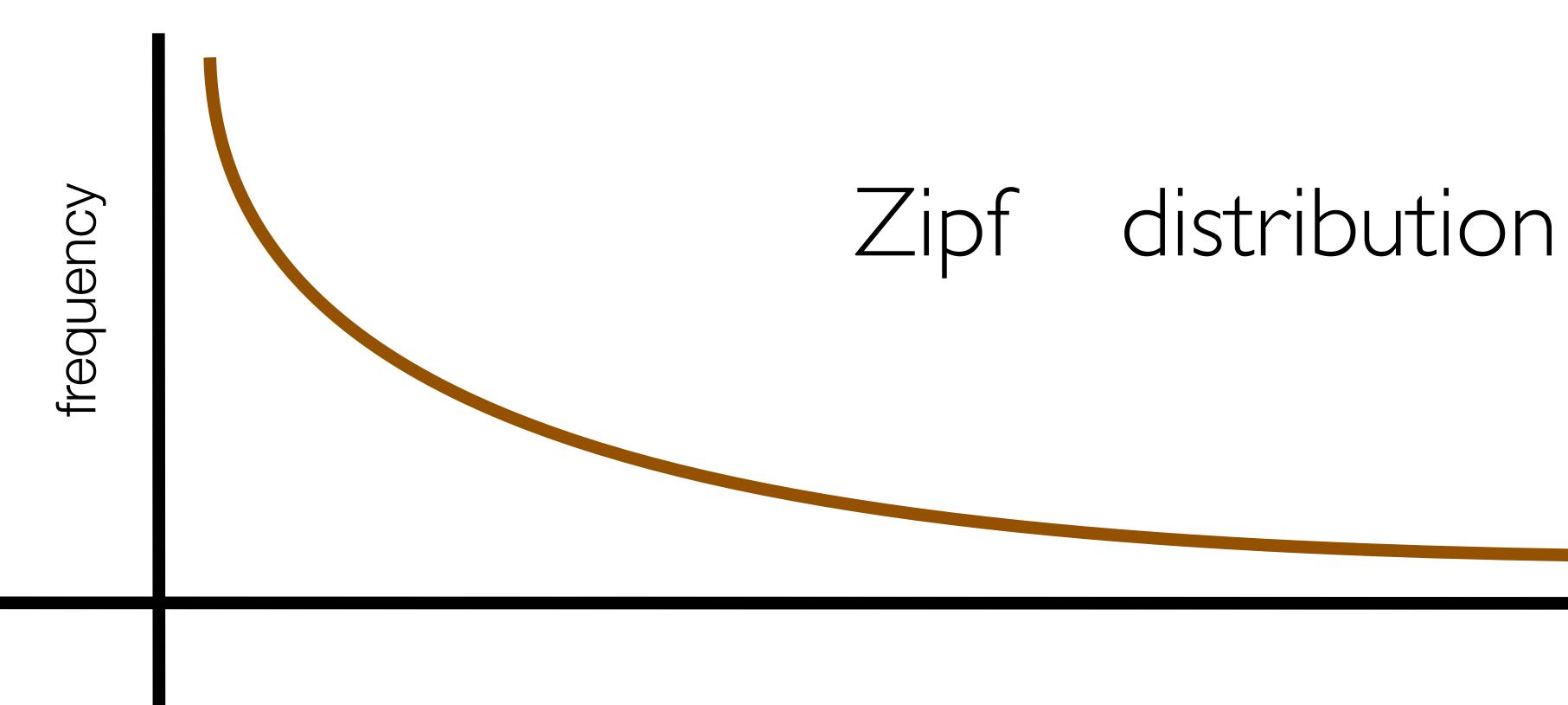
- We would like a database of speech which contains
 - every possible speech base unit type
 - in every possible context
- This list of desired base unit types in context will be very, very long
 - if we limit the scope of context, the list will be finite
- Will it be possible to record one example of every unit-in-context?

Large number of rare events

- A few types (units-in-context) are very frequent
- A large number of types are individually very infrequent
 - but the large number of such types means that together they make up a significant proportion of (spoken) language
- to synthesise

• There is a high chance that we will need **at least one** rare type in **any sentence** we have





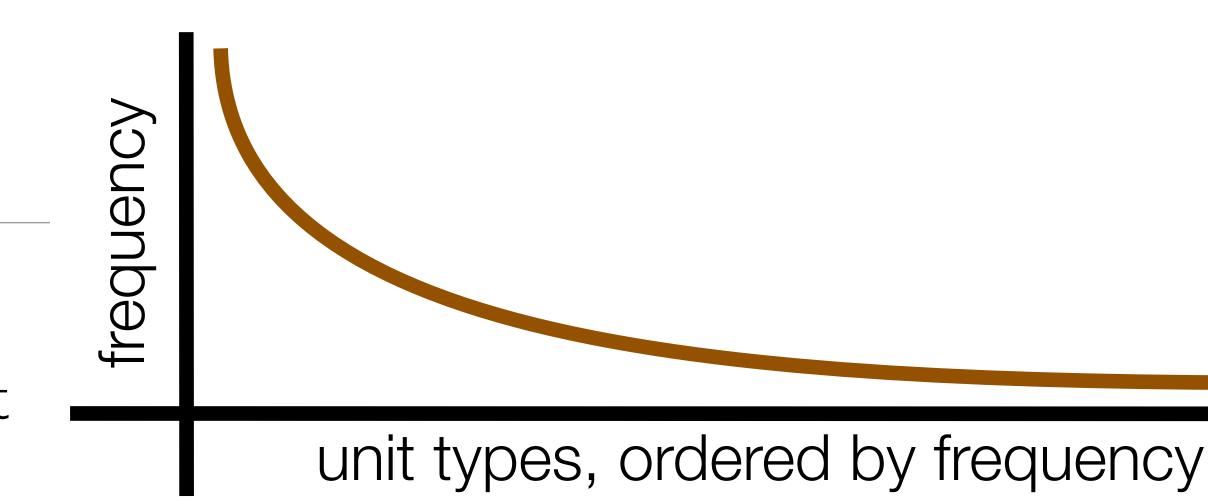
unit types, ordered by frequency

Databases for speech synthesis

- key concepts
- <u>script design</u>
- annotating the database

Why design a script?

- In randomly-chosen natural text
 - Zipf-like distribution of units-in-context
- As database size increases
 - number of tokens of frequent types increases rapidly
 - number of infrequent types with at least one example grows very slowly
 - many (most!) types will have **no tokens** at all, even for very large database sizes
- In practice, it will be impossible to find a set of sentences that includes at least one token of every unit-in-context type
- So, try to **design** a script that is better than random selection of sentences





Goals of script design

- Cover as many types (in context) as possible
 - increase chance of finding an exact match at synthesis time although still very unlikely
 - maximise the variety of contexts in which each base unit type occurs
 - the target cost will differentiate between them
 - join cost has better chance of finding unit sequences that concatenate well
- With as **few tokens** as possible i.e., in as few sentences as possible
 - recording speech is time consuming
 - harder to maintain consistency over longer recording periods (days, weeks, months) • in unit selection, the run-time system will include a copy of the database

Typical approach to script design: a greedy algorithm for text selection

1. Find a very large text corpus

- e.g., newspaper text, out-of-copyright novels, web scraping 2. Make an exhaustive 'wish list' of all possible **types** (in context) that we would like **3.** Find the sentence in the corpus which provides the largest number of different **types**
- that we don't already have
- **4.** Add that sentence our recording script
- 5. Remove those types from the 'wish list'
- 6. If recording script is long enough, stop. Otherwise, go to 3.

Where do we get this "very large text corpus"?

- Out of copyright literature (old novels)
 - e.g., as used in the ARCTIC corpora
- Newspaper text
 - usually copyrighted, so must obtain permission to use
- Problems with most sources of text:
 - written text is not usually intended to be read aloud
 - prosodic variation will therefore be limited
 - long sentences lead to insufficient phrase initial/final segments

Example of text selection

- We'll assume that we have a large corpus of text to start from
- Corpus cleaning

 - Discard all sentences that contain out-of-vocabulary (OOV) words
 - Discard all sentences that are too long (hard to read out loud) or too short (atypical prosody)
 - Optional: discard hard-to-read sentences
- Front-end processing
 - Pass the text through the TTS front end to obtain, for each sentence
 - base unit sequence (e.g., diphones)
 - linguistic context of each unit (e.g., stress)

• Define the vocabulary (e.g., only words in our dictionary, or the most frequent words in the corpus)

The wish list

- Define the base unit **type** let's use diphones in this example
- Which would give this wish list:

aa_aa aa_f aa_ae aa_g aa_ah aa_hh aa_ao aa_ih aa_aw aa_iy aa_ay aa_jh aa_b aa_k aa_ch aa_l aa_d aa_m aa_dh aa n aa_eh aa_ng aa_er aa_ow aa_ey aa_oy

The wish list

- In reality we want every type in every context
- What context? let's just consider stress
- Which would give this wish list:

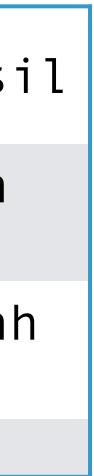
aa_aa unstressed aa aa stressed aa ae unstressed aa_ae_stressed aa_ah unstressed aa ah stressed aa ao unstressed aa ao stressed aa aw unstressed aa aw stressed aa_ay_unstressed aa_ay_stressed aa b unstressed

aa b stressed aa ch unstressed aa ch stressed aa d unstressed aa d stressed aa dh unstressed aa dh stressed aa eh unstressed aa_eh stressed aa er unstressed aa_er stressed aa ey unstressed aa_ey_stressed

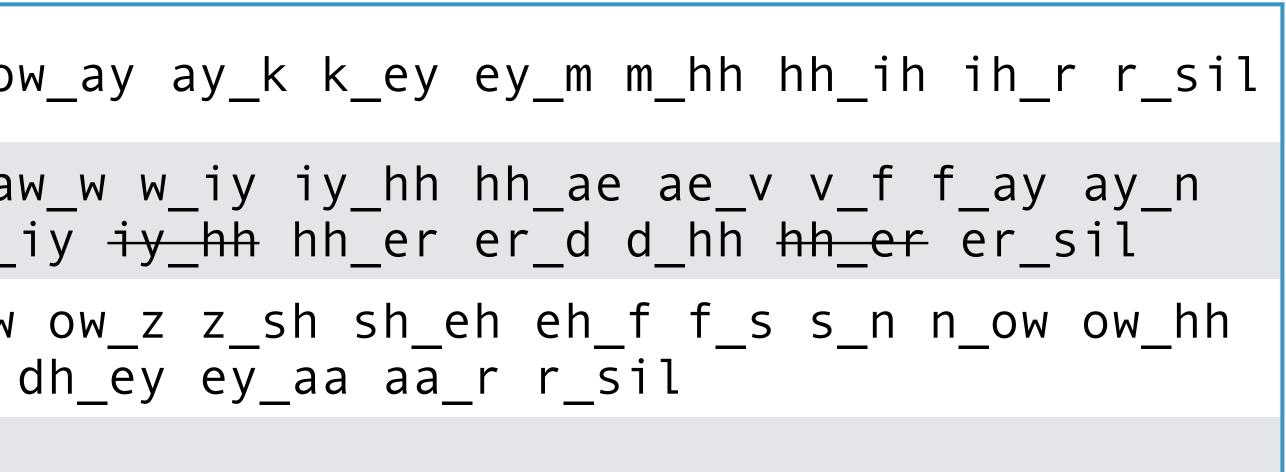
 $\bullet \bullet \bullet \bullet \text{etc.}$

Create an index of all available sentences, and the units they contain

So I came here.	sil_s s_ow c
Now we have finally heard her	sil_n n_aw a n_ax ax_l l_
Those chefs know who they are.	sil_dh_dh_ov hh_uw_uw_dh
etc	



Select	richest sente	nce. Mc	ove it to the sc	ript. Upda	ate wish list	•	
Solca	ame here.	sil_s	s_ow ow_ay ay_	k k_ey ey_m m_	_hh hh_ih ił	ı_r r_s	
Now we have finally heard her. <pre>sil_n n_aw aw_w w_iy iy_hh hh_ae ae_v v_f f_ay ay_n Now we have finally heard her. n_ax ax_l l_iy iy_hh hh_er er_d d_hh hh_er er_sil</pre>							
Those chefs know who they sil_dh dh_ow ow_z z_sh sh_eh eh_f f_s s_n n_ow ow_hl are. hh_uw uw_dh dh_ey ey_aa aa_r r_sil							
etc							
aa_aa aa_ae aa_ah aa_ao aa_aw aa_ay aa_b aa_b aa_b aa_ch aa_ch aa_eh aa_en aa_er aa_ey	aa_f aa_g aa_hh aa_ih aa_iy aa_jh aa_jh aa_k aa_l aa_n aa_n aa_ng aa_ow aa_oy	ay_ey ay_f ay_g ay_hh ay_ih ay_iy ay_jh ay_jh ay_k ay_l ay_n ay_n ay_ng ay_ow	ey_f ey_g ey_hh ey_ih ey_iy ey_jh ey_jh ey_k ey_l ey_n ey_n ey_ng ey_ow ey_ow	<pre>hh_f hh_g hh_hh hh_ih hh_iy hh_jh hh_l hh_l hh_l hh_n hh_n hh_n hh_n hh_</pre>	zh_f zh_g zh_hh zh_ih zh_iy zh_jh zh_j zh_k zh_l zh_n zh_n zh_ng zh_ow zh_oy	zh_p zh_r zh_s zh_sh zh_t zh_t zh_u zh_u zh_u zh_v zh_v zh_y zh_z zh_z	



Optional improvements

- Guarantee **at least one token** of every base unit type
- Try to cover the **rarest** units first
 - more common units will be selected anyway, as a by-product
- How to define "rarest"?
 - count occurrences in the original large corpus
- How to implement this
 - to their frequency

• include weights in the "richness" measure that reward rarer units in inverse proportion

Optional: domain-specific script

- **1.** Select (or manually design, or automatically generate) in-domain sentences
- 2. Measure coverage obtained so far
- **3.** Fill in the gaps in coverage, using sentences selected from the large text corpus

Databases for speech synthesis

- key concepts
- script design
- <u>annotating the database</u>

Orientation

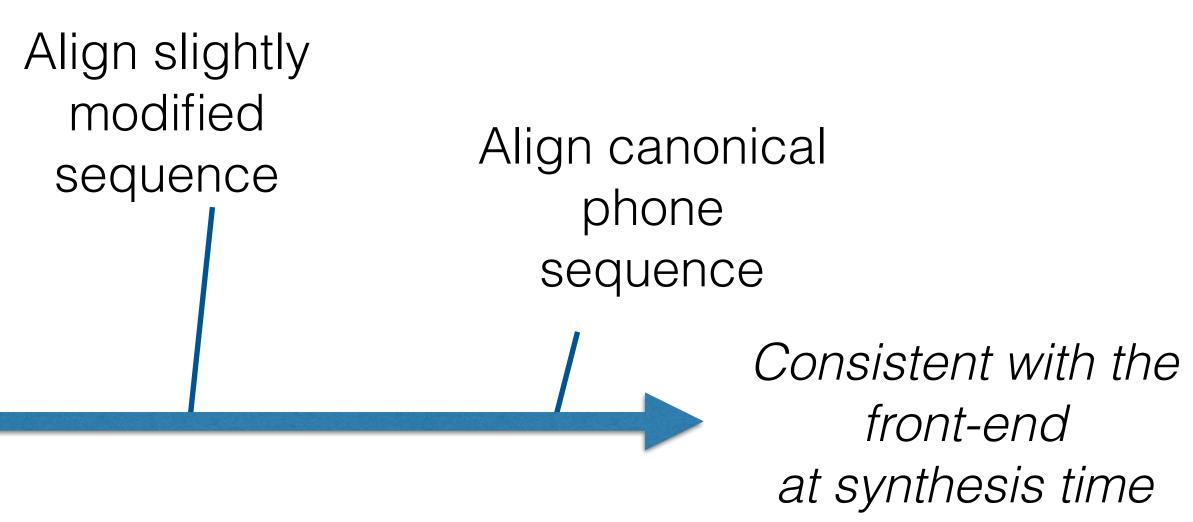
- What have we got?
 - a script composed of sentences
 - a recording of each sentence
- What remains to be done?
- a time-aligned phonetic transcription of the speech
- annotate the speech with supra-segmental linguistic information



Why not simply hand-label the speech ?

Hand-label from scratch

Faithful to how the speaker spoke





"The text sentence that the speaker read out"

Analytical labelling

- Two reasons we prefer analytical labelling to be done automatically
 - more <u>consistent</u>
 - the labels are just an index to retrieve units from the database
 - we want the predictions from text to match the labels on the database
 - faster and <u>cheaper</u>
- manual correction of automatic labels
 - standard practice in some commercial systems
 - mainly this involves fixing gross errors such as mis-alignment

Forced alignment

- Standard technique from Automatic Speech Recognition
- The same as full-blown speech recognition, except
 - we have a very highly constrained language model
 - because we know the word sequence
 - during decoding, we record the model- (or state-) level alignment

Ingredients for forced alignment

- <u>Acoustic models</u>
 - a fully-trained set of phone models
- Pronunciation model
 - the same dictionary we will use for synthesis
 - can include pronunciation variation
 - plus optional rule-based variations, such as vowel reduction
- Language model
 - constructed from the known word sequence for the current sentence
 - i.e., language model is different for each sentence
 - can insert optional silences between words

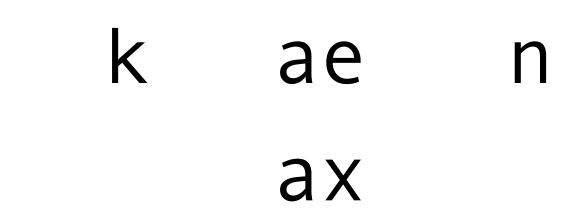
Language model

there was a



Pronunciation model = dictionary + optional vowel reduction

"...what can it do for..."





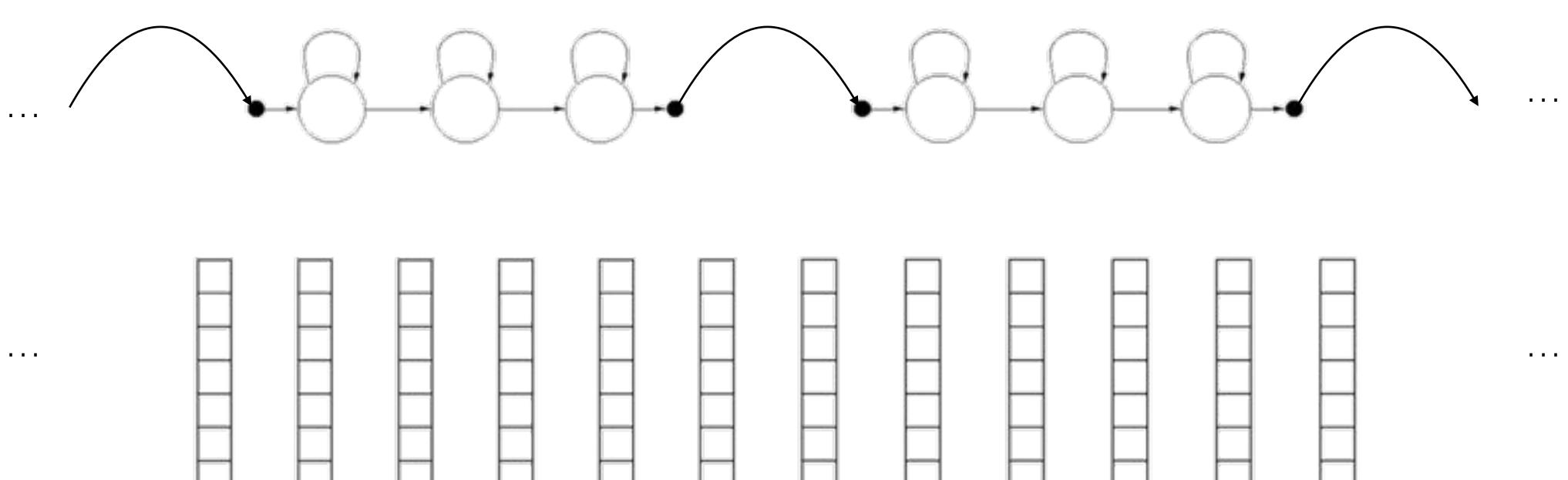
Acoustic model

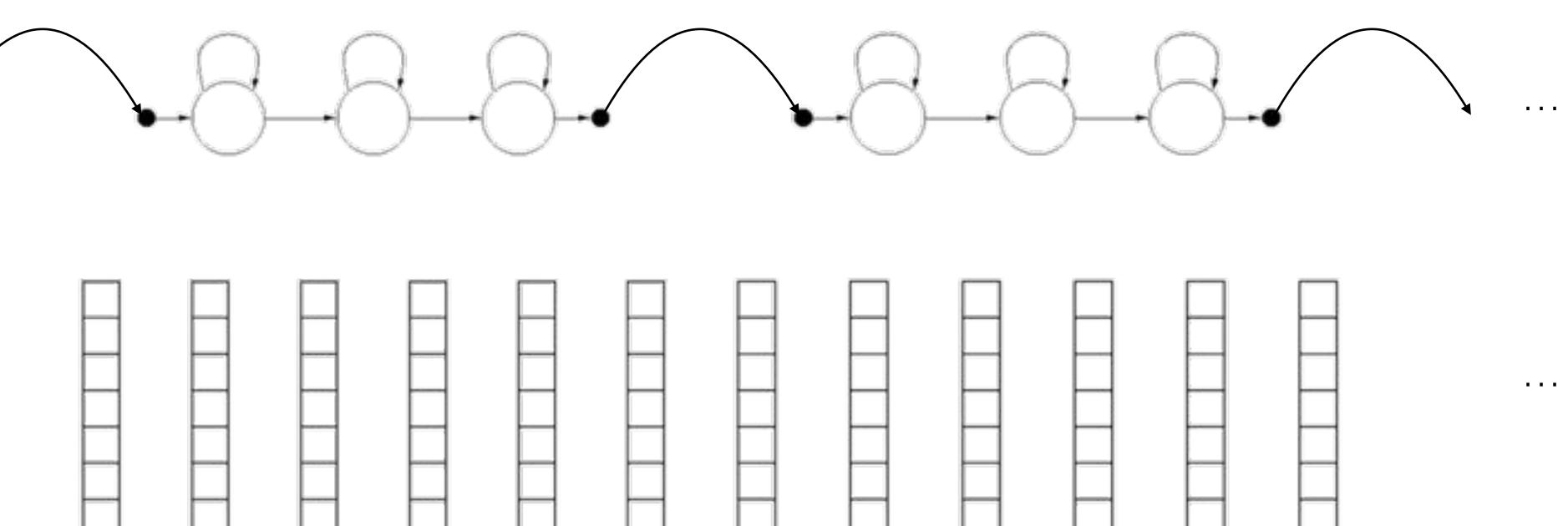
- Could borrow from an existing ASR system
 - actually tend to get better results with simpler, speaker-dependent models
 - trained on the speech database
- Hang on: training on the "test data"? Isn't that cheating?
 - no because it's not "test data"!

Training an acoustic model on the recorded speech data

- We only have word transcriptions
- Aligned at sentence boundaries
- We already know about basic Automatic Speech Recognition
- training models on data with model-level (e.g., whole word) aligned transcriptions • But, even there, we did not need state-level alignments
- Can generalise this to not needing model-level alignments
 - concatenate models, to make an acoustic model for a particular whole sentence
 - this is just a single (albeit rather long) HMM, and we know how to train that

Flat start training of HMMs





Optional silences between words



An acoustic model of optional silences, to use between words

- Single-state model
- Skip transition allows model to emit a sequence of **zero or more** observations

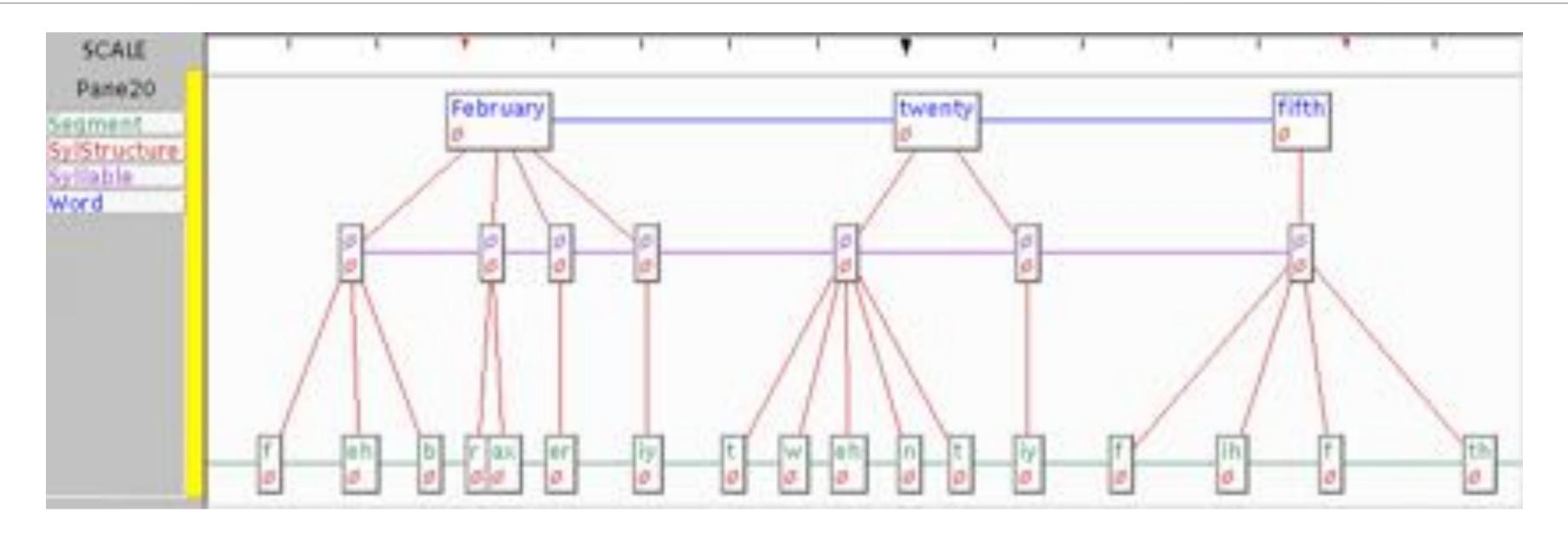
• Emission parameters (e.g., a GMM) are tied to the centre state of the long silence model

Compile language model + pronunciation model + acoustic model

there was a change sil ax sp ax sp sp ax

now dh eh r w aa z ax ch ey n jh n aw sil Sp ах

Combining aligned phone sequence with supra-segmental structure



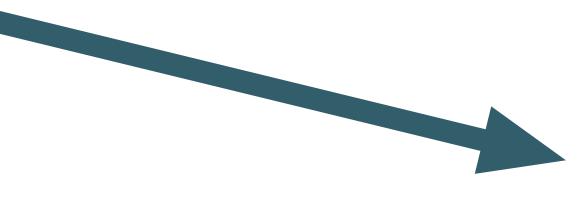
sil fehbraxeriysptwehntiyf ih f th sil 11.1 atta Ultranata La Strictly out of the light of the light of the



- How good is our synthetic voice?
- It's time to evaluate it, but how?
 - listen to it ourselves?
 - ask others to listen to it?
 - measure objectively?
- What precisely do we want to measure? Why?



- How good is our synthetic voice?
- It's time to evaluate it, but how?
 - listen to it ourselves?
 - ask others to listen to it?
 - measure objectively?
- What precisely do we want to measure? Why?



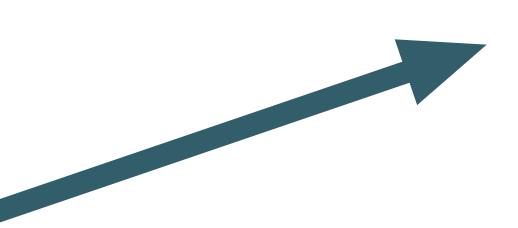
Can we judge that in isolation, or must it be in comparison to another system/voice?

- How good is our synthetic voice?
- It's time to evaluate it, but how?
 - listen to it ourselves?
 - ask others to listen to it?
 - measure objectively?
- What precisely do we want to measure? Why?



Each of these has advantages and disadvantages that we need to consider.

- How good is our synthetic voice?
- It's time to evaluate it, but how?
 - listen to it ourselves?
- ask others to listen to it?
- measure objectively?
- What precisely do we want to measure? Why?



Naturalness? Intelligibility? Something else?