

3. Singer Library

The Singer Library (Library hereafter) is a database of samples (mostly diphones) extracted from real people's singing.

The samples must include all possible combinations of phonemes of the target language. In English case, all possible combinations of C-V, V-C, V-V are recorded, processed, and put into the Library. The developer can optionally add polyphones with more than two phonemes.

Sustained vowels are also put into the Library. They are used to reproduce the behavior of sustained vowels, which are essential in singing synthesis.

The number of samples is approximately 2000 per one pitch. In order to record these samples effectively, a special script is designed. After a recording is done, the recorded wave files are semi-automatically segmented and necessary segments are extracted.

The Library used for the song we submitted includes samples in three pitch ranges.

4. Synthesis Engine

The Synthesis Engine receives score information, selects necessary samples from Singer Library and concatenates them. In Figure 4, a block diagram of synthesis engine is shown.

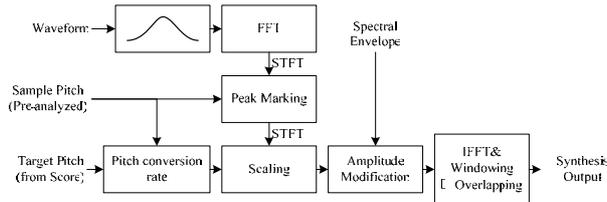


Figure 4: Block Diagram of Synthesis Engine

The problem in concatenating samples is that the samples are recorded in different pitches and different phonetic contexts, i.e. the pitch must be converted to a desired one, and the timbre must be "smoothed" around the junction of samples. In the synthesis engine, the pitch conversion and the timbre manipulation are done in frequency domain.

The pitch conversion is done by "scaling" spectrum. After getting STFT of a sample waveform, the power spectrum is divided into several regions. For each region, the spectrum is scaled so that the scaling factor corresponds to the pitch conversion. The local spectral shape near each harmonic is kept as it is (hence non-linear scaling). The timbre manipulation is done by changing an amplitude of each harmonic.

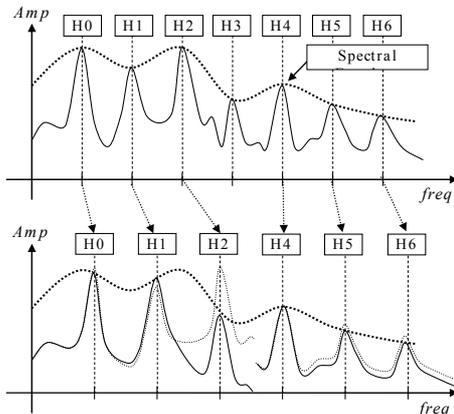


Figure 5: Pitch Conversion and Timbre Manipulation

In changing the pitch, the phase needs to be corrected. Assuming perfect harmonic, the following compensation value is added to the phase for i th harmonic.

$$\Delta\varphi_i = 2\pi f_0 (i+1)(T-1)\Delta t \quad (1)$$

where T is a pitch conversion ratio of f_{0T} (after conversion) and f_0 (original pitch), Δt is a frame duration.

The timbre of a sustained vowel is generated by interpolating spectral envelopes of the surrounding samples. For example, if you would like to concatenate a sequence s-e, e, e-t (which is a part of [set]), the spectral envelope of sustained [e] at each frame is generated by interpolating [e] in the end of [s-e] and [e] in the beginning of [e-t]. By doing these processes, there is theoretically no timbre gap in concatenation.

Sample timing is automatically arranged so that the vowel onset of a syllable should be strictly on the "Note-On" position.

5. Acknowledgements

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6. References

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