

Signal Reconstruction from Sampled Data Using Neural Network

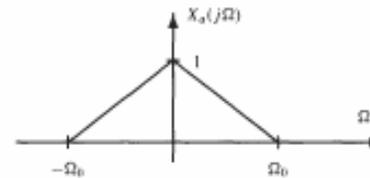


Poster presentation by
Ron Teichner and Asya Leikin

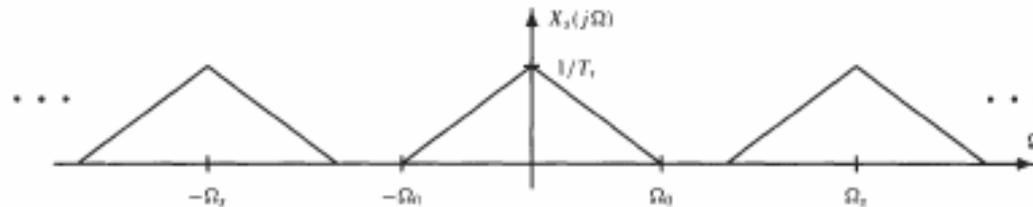
Supervisor: Dorit Baras

Sampling and Reconstruction

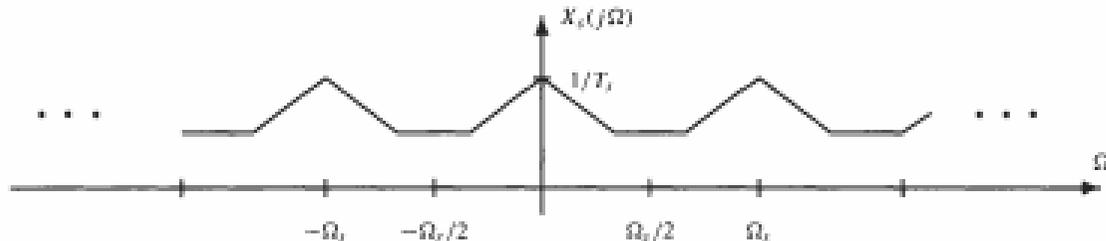
For the following signal:



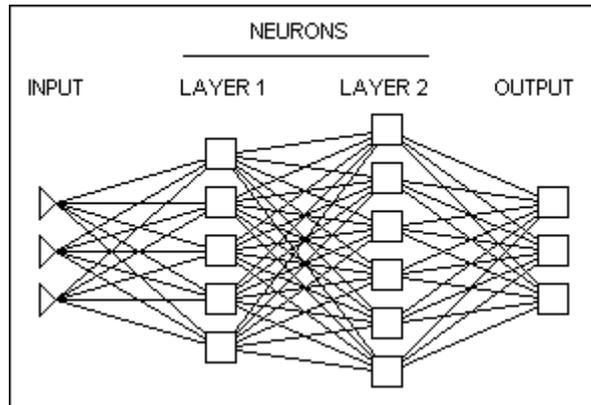
- Sampling above Nyquist sampling rate:



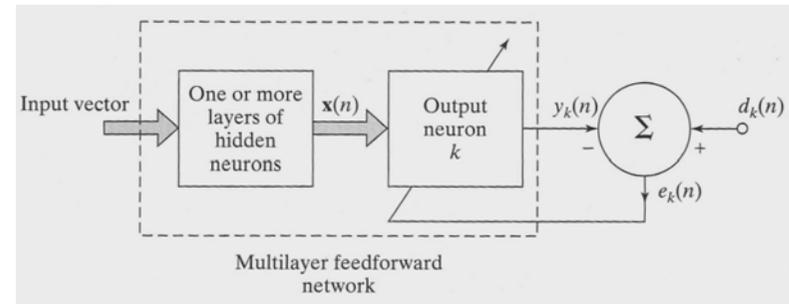
- Sampling below Nyquist sampling rate – aliasing:



Neural Network



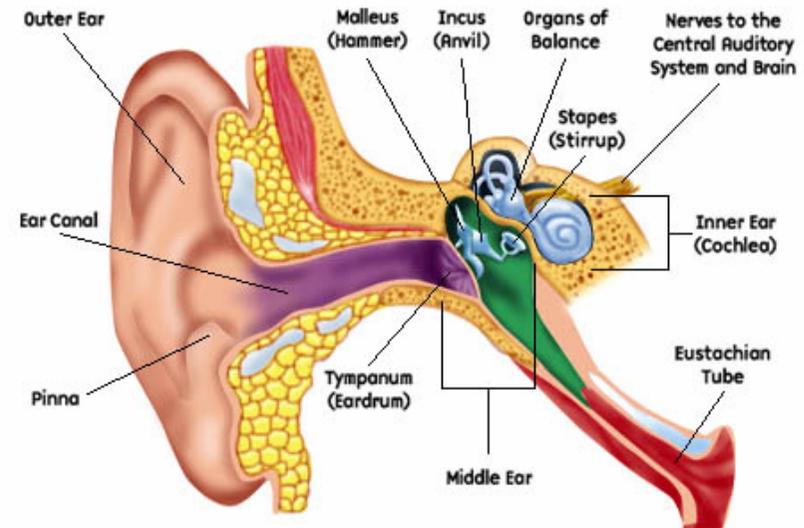
A Neural Network generally maps a set of inputs to a set of outputs after a training procedure on a finite set of input-output examples.



Training is the act of presenting the network with some sampled data and modifying the weights to better approximate the desired output

The Human auditory system

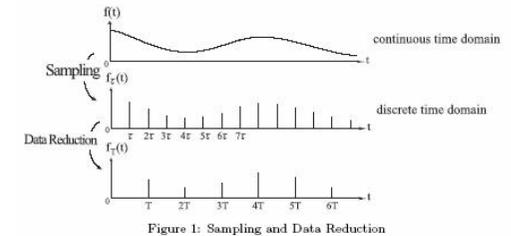
- A very complex system with the phenomena of:
 - Sensitivity to log-spectrum - Octaves
 - Critical bands
 - Threshold of hearing
 - Masking



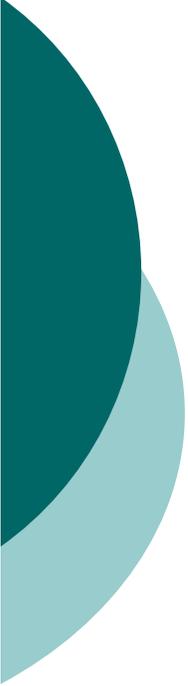
The Paper:

“SIGNAL RECONSTRUCTION FROM SAMPLED DATA USING NEURAL NETWORK”

Akihito Sudou, Pitoyo Hartono, Ryo Saegusa, Shuji Hashimoto



- A signal sampled with Nyquist rate underwent data reduction.
- Reconstruction was made both with a linear perceptron and a multilayered perceptron (with one hidden layer).
- For comparison a reconstruction using FIR filter with sinc was made.
- The NN did the training by minimizing the MSE.
- Two experiments were made:
 - Comparing the errors between the different reconstructions and the original signal for different data reduction rates.
 - Comparing the results of the different reconstructions while changing the number of inputs to the NN and the data reduction rate.



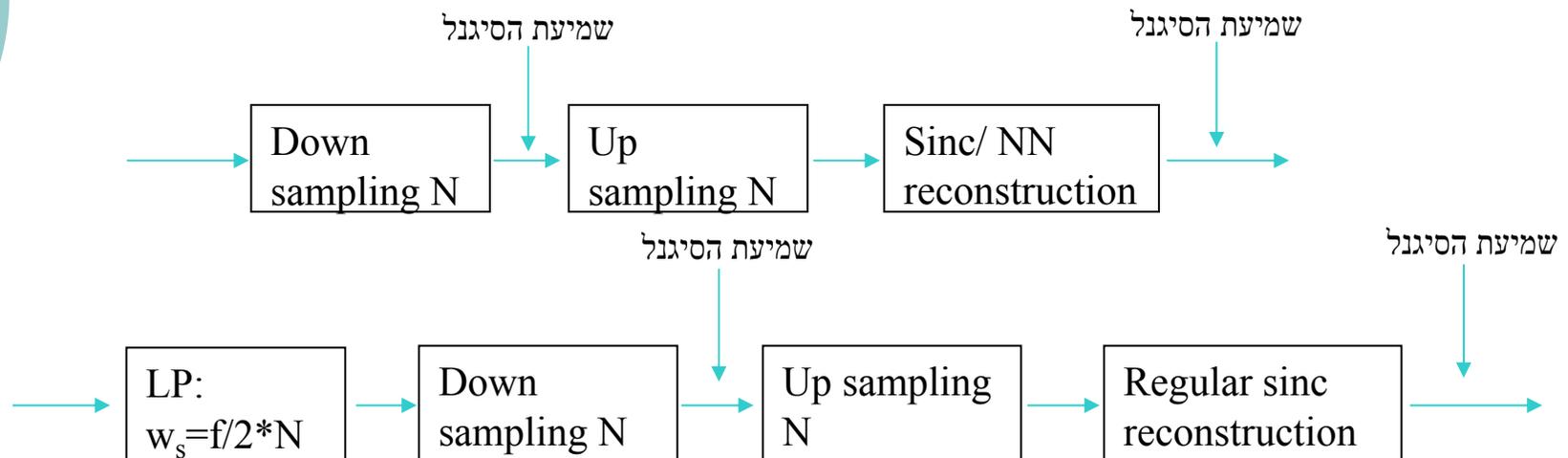
Results from the paper:

Note: the criteria for a good reconstruction was the average MSE not the sound quality.

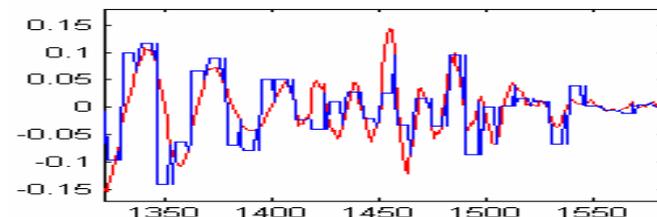
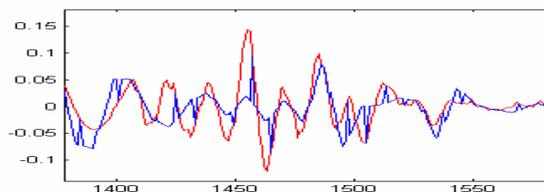
- The experiment showed that the perceptron achieved better reconstruction than the FIR reconstruction. Also it was shown that for larger data reduction the mistake in the reconstruction is larger.
- They didn't see a meaningful difference in the reconstructions by the linear perceptron and the multi-layer perceptron with one hidden layer.
- The second experiment showed that using a perceptron with larger number of inputs improves the reconstruction

Time Domain

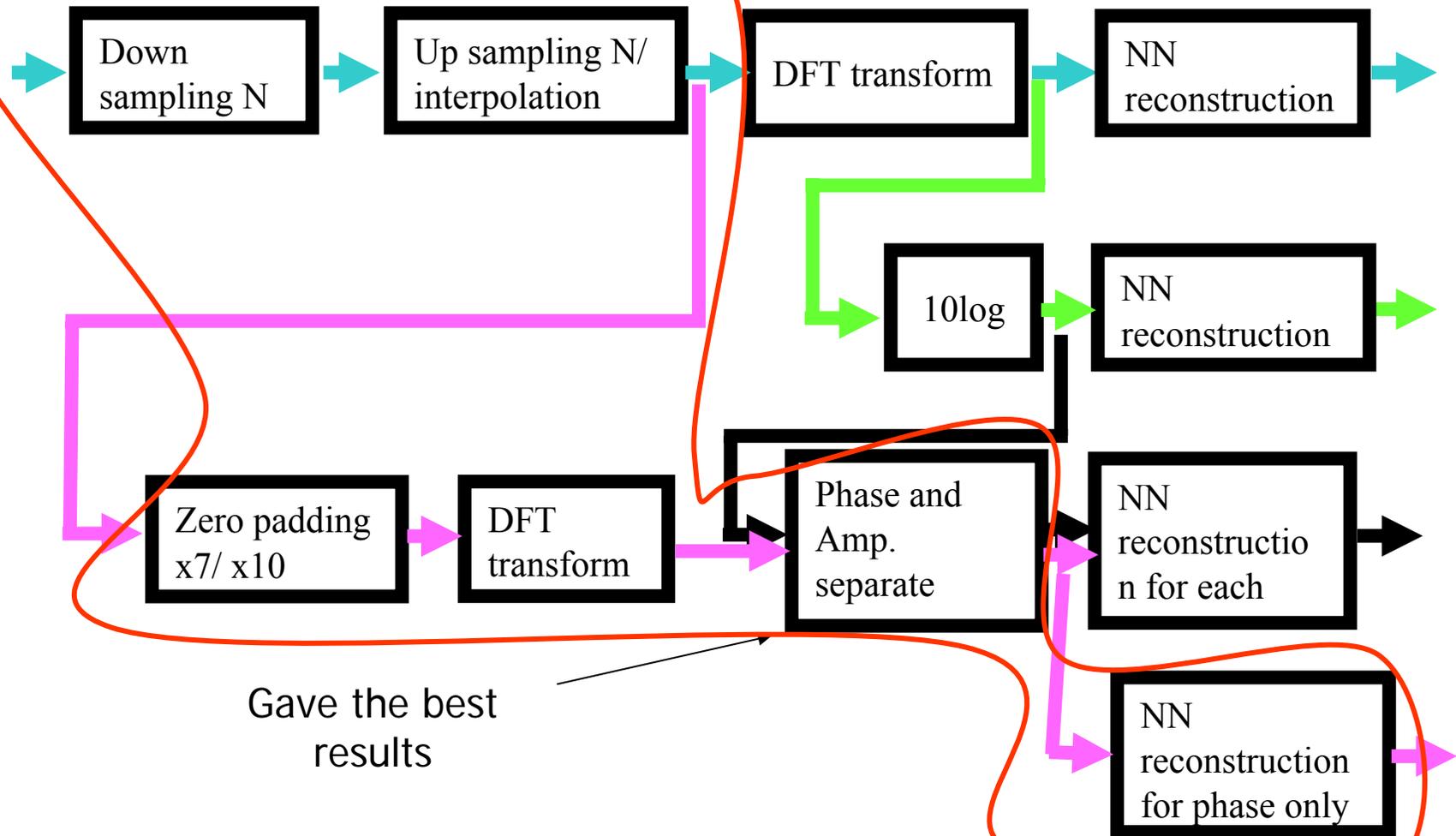
- We are using 2 signals: speech and musical.



- up sampling does zero interpolation and is probably not optimal for the reconstruction. We tried other types of interpolation- linear: nearest neighbor:



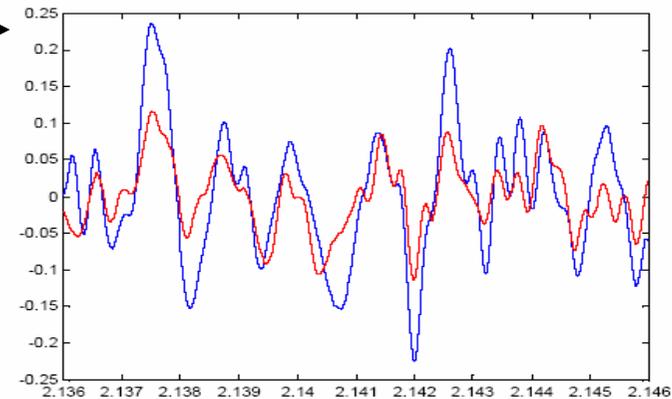
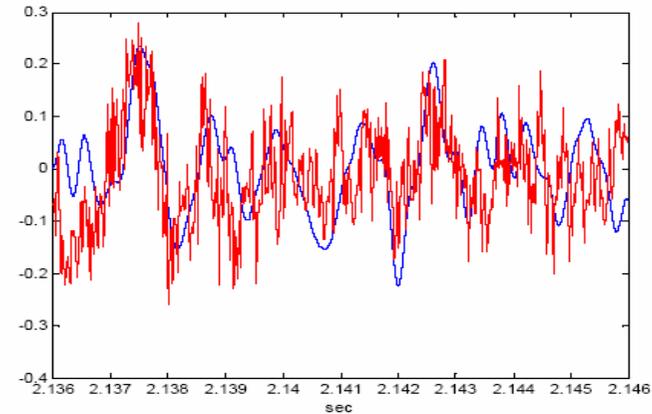
Frequency Domain



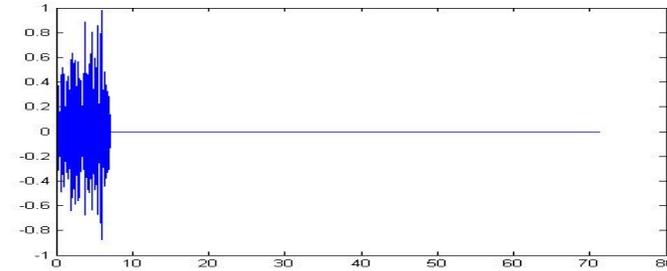
Gave the best results

Phase and amplitude reconstruction vs. only phase reconstruction:

- Red- the reconstructed signal
- Phase and amplitude reconstruction and then combination →
- Using the unreconstructed amplitude and the reconstructed phase.
- We can see that the amplitude reconstruction adds a lot of noise.



Zero padding



- Zero padding is actually a way to increase the frequency resolution.
- We did only phase reconstruction and used either the original (non-reconstructed amplitude) or the average amplitude.
- The average amplitude showed better results
- We used PESQ and our hearing as an estimate for the quality of reconstruction here are the best results that we got for the speech signal.

| | PESQ | Comp. rate |
|------------------------|-------|------------|
| L=30 segl=0.1 pad=7 | 1.679 | 1.276 |
| L=30 segl=0.1 pad=6 | 1.677 | 1.413 |
| L=10 segl=0.1 pad=7 | 1.670 | 2.305 |
| L=20 segl=0.1 pad=7 | 1.685 | 1.643 |
| L=30 segl=1 pad=7 | 1.716 | 3.297 |
| L=30 segl=0.5 pad=7 | 1.689 | 2.803 |

Sending and receiving:

○ Sending:

Down sampling,
interpolation,
Zero padding



FFT transform,
10log, phase
and amplitude
separation



NN training on
the phase and
saving the
weights



Sending: I- length of the signal after down sampling, N- down sampling rate, type of the interpolation, zero padding rate, L- window length, segment length, down sampled signal, weights in the correct order.

○ Receiving:

Interpolation,
Zero padding



FFT transform,
10log, phase
and amplitude
separation



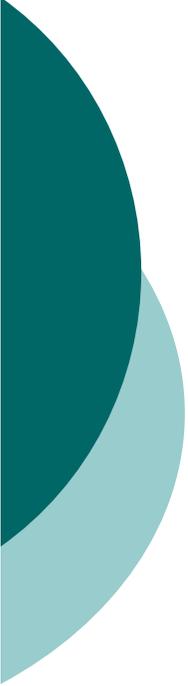
Phase
reconstruction using
the weights
received, finding the
average amplitude



Composing
the signal
from the
reconstructed
phase and the
average amp.



Noise reduction in
time domain



Conclusions

- The reconstructed signals are very noisy and need a lot of noise reduction processing if a good quality of sound is required but if the goal is only to understand the sentence spoken the quality is good enough.
- This method proved to be not useful as a compression method because there are much better ways of compression that deliver a much better sound quality after decompression.