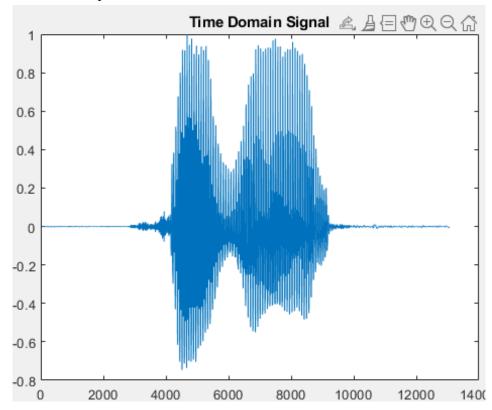
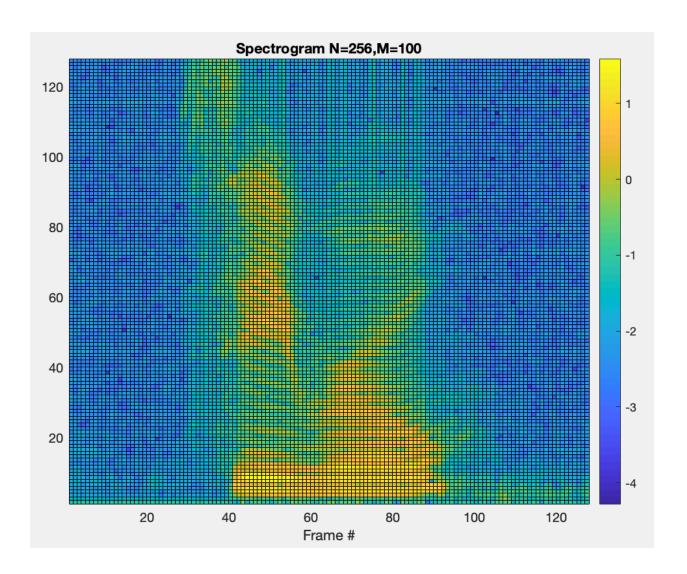
Final Project Tests

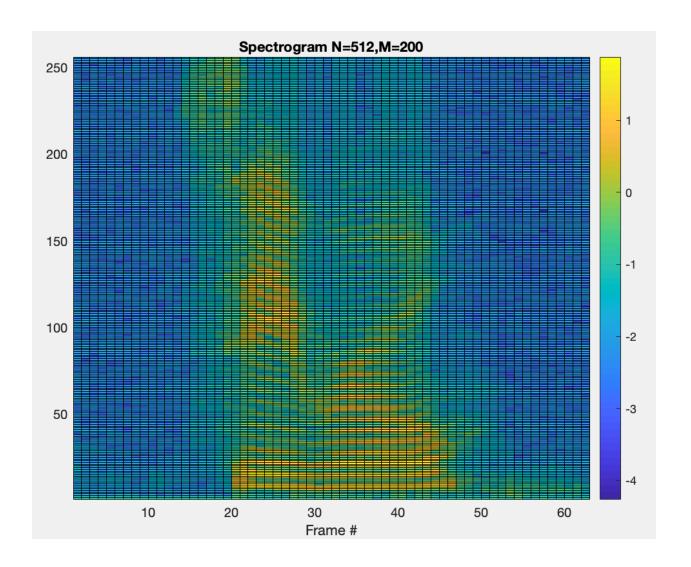
Test 1:

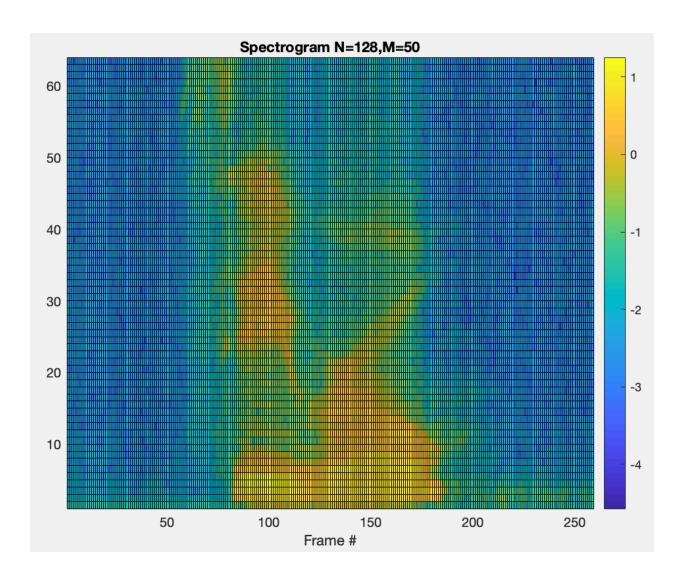
Human accuracy = 75%

Test 2: Sampling rate = 12.5 KHz Time = 20.48 ms per frame

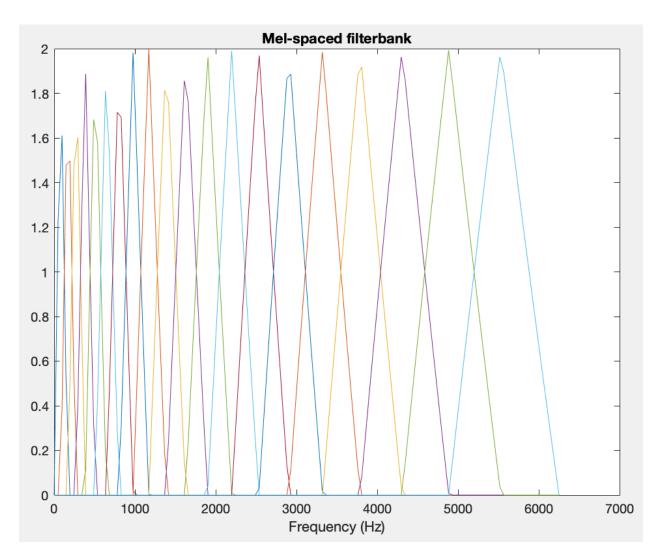


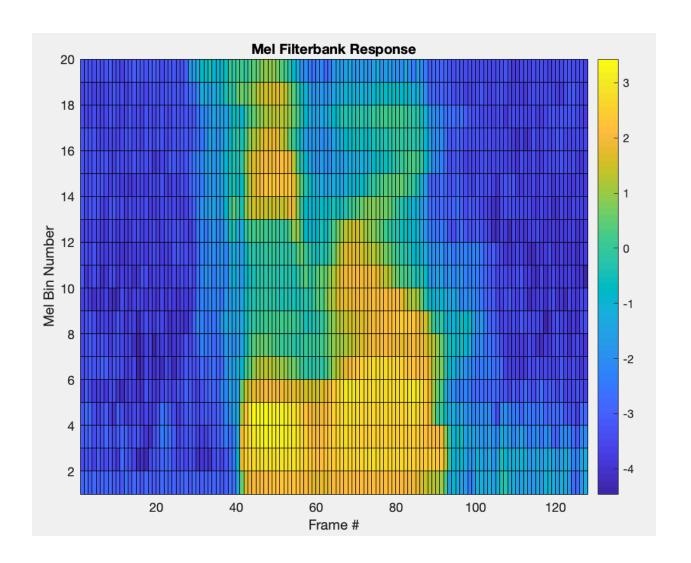




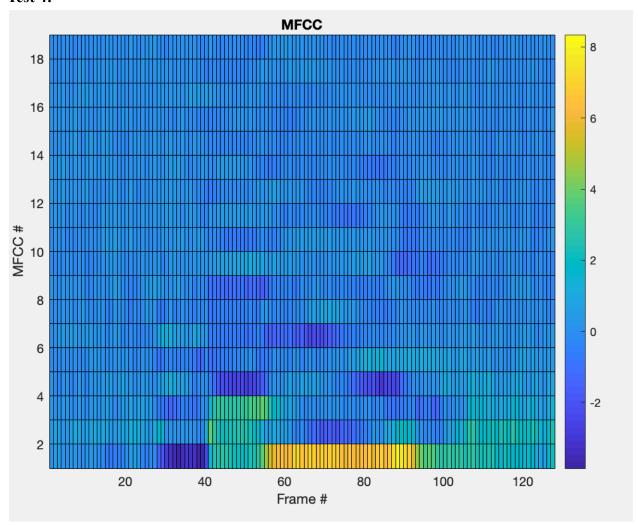


Test 3: Mel filterbank plot looks somewhat jagged due to the discretization error of having only N = 256 points. They are not all perfectly triangular and they don't peak at the same amplitudes.

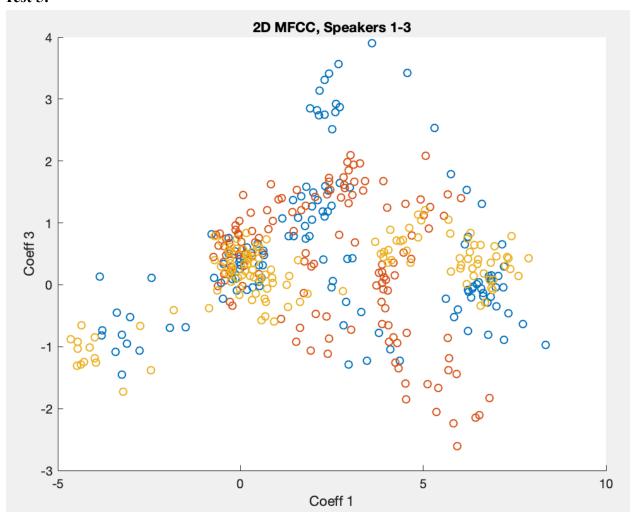




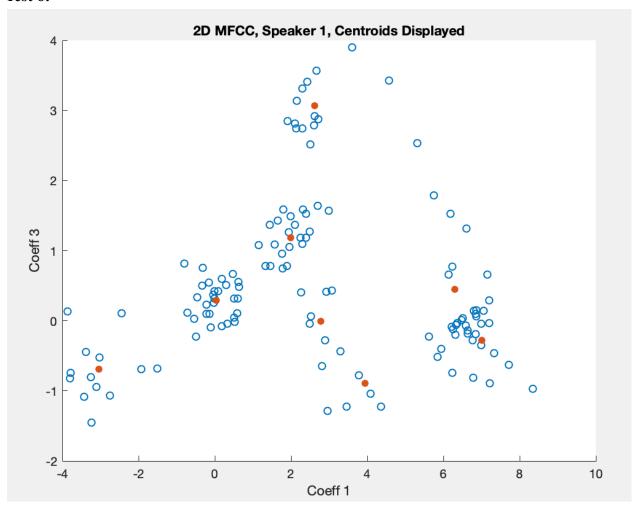
Test 4:



Test 5:



Test 6:



Test 7: With parameters, 256,100,20,.001,.01, and 8 for N,M,P, thresh, epsilon and codeword_lim respectively, code gets 100% correct on original test audio, much better than human 75% correct

Test 8:

Reduction in accuracy due to notch filtering is highly dependent on which frequency the notch filters out, the lower frequencies where most of the speech energy is have a much larger effect. With notch at .5*pi, accuracy is reduced to 87.5% on original audio.

Test 9: 100% accuracy on group member's voices

Test 10:

82.14% overall accuracy on 28 test samples, and 16 different training voices