

Attentive Filtering Networks for Audio Replay Attack Detection

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Research Problem & Our Objectives

Problem

Automatic speaker verification (ASV) systems are susceptible to malicious spoofing attacks, especially those in the form of audio replay.



(Left) An example of audio replay attack [1]. The left phone (black color) is a smart phone with a voice-unlock function for user authentication; the right phone (white color) replays a pre-recorded speech sample to unlock the left phone.

Objectives

1. Advance previous anti-spoofing research on DNN based system, and
2. Develop a system that automatically acquires and enhances discriminative features in both the time and frequency domain.

ASVspoof 2017 Version 2.0

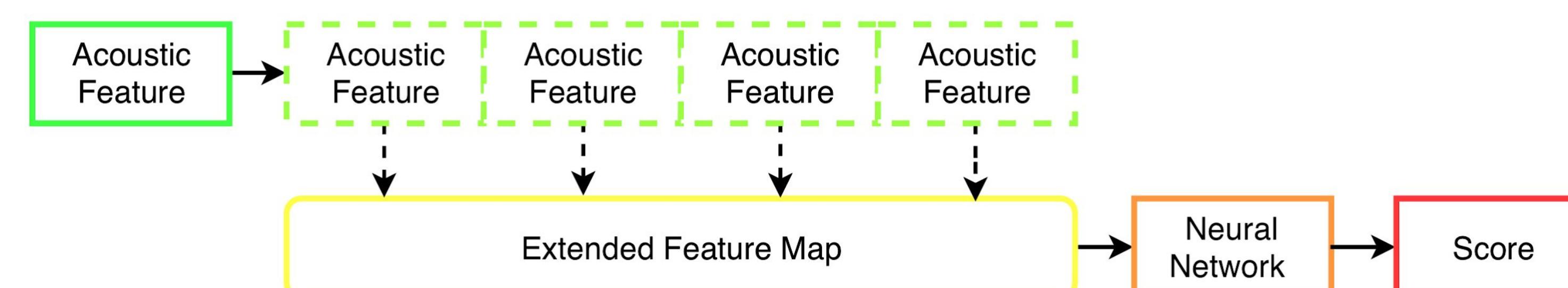
The ASVspoof 2017 corpus is a collection of *bona fide* and spoofed utterances. *Bona fide* utterances are a subset of the RedDots corpus, while the spoofed utterances are the result of replaying and recording *bona fide* utterances using a variety of heterogeneous devices and acoustic environments [2].

Subset	# Spk	# Replay sessions	# Replay Config	# Utterances
	Bona fide	Replay		
Training	10	6	3	1507 1507
Devel.	8	10	10	760 950
Eval.	24	161	57	1298 12008
Total	42	177	61	3565 14465

Unified Feature Map Creation

Acoustic Feature

Log magnitude spectrum (logpsec) is used as the acoustic feature. We kept all frames without applying VAD and applied mean normalization using a 3-s sliding window.



Extended Feature Map

The unified time-frequency map was created by extending all utterances to the length of the longest utterance by repeating their feature maps. The benefits of this feature engineering approach is that there is no need for feature truncation or frame-level score combination.

Code & Contact

Code: github.com/jefflai108/Attentive-Filtering-Network
Alternatively, you can reach the author at clai24@mit.edu

Attentive Filtering Network

Attentive Filtering

Attentive Filtering (AF) accumulates features in frequency and time domains selectively. AF augments every input feature map S with an attention heatmap A_s to produce a new feature map S^* for the DRN. Mathematically,

$$S^* = A_s \circ S + \bar{S}$$

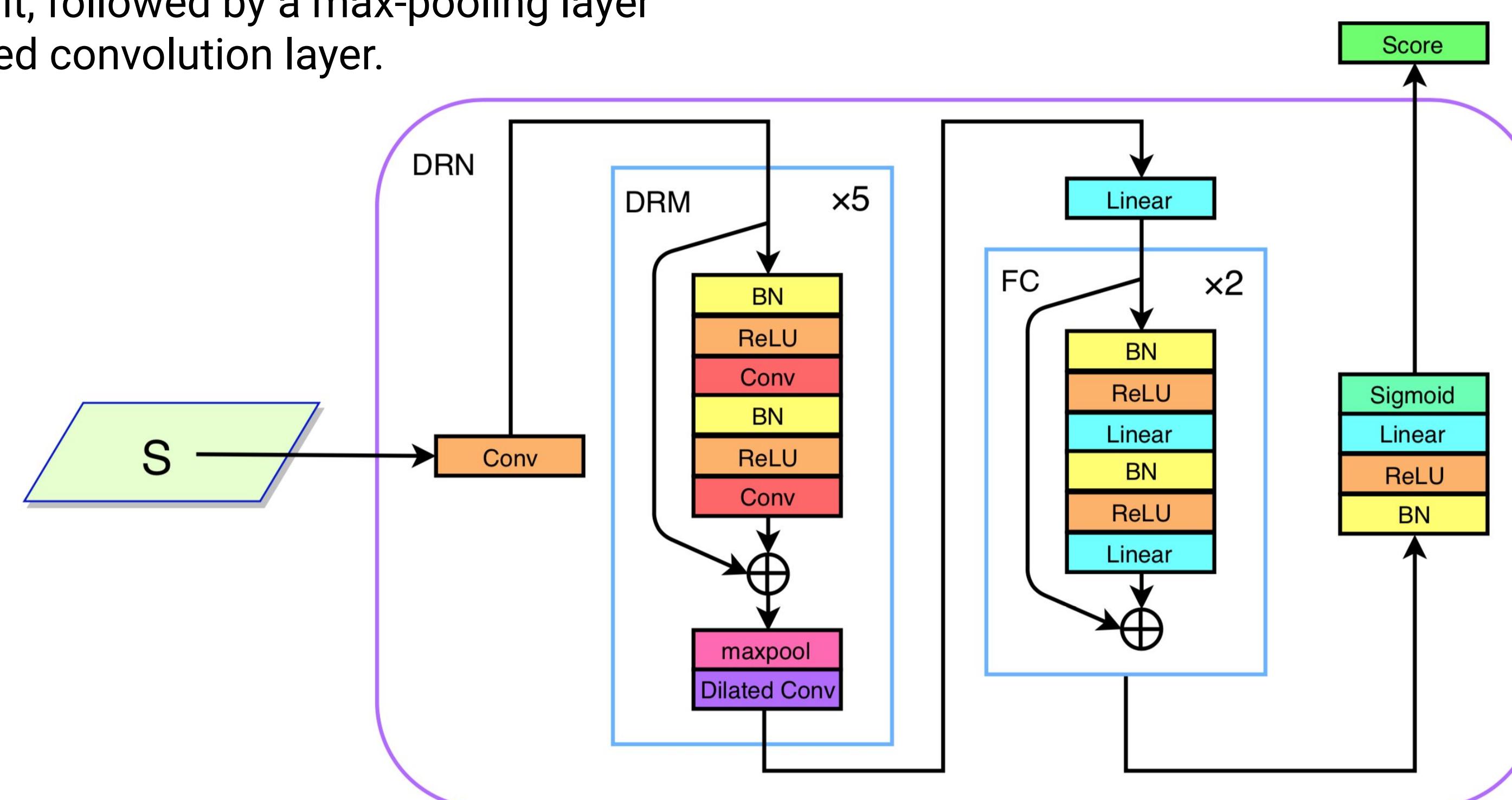
We set \bar{S} as the residual S . Attention heatmap A_s is described as,

$$A_s = \phi(U(S))$$

ϕ is a nonlinear transform such as *sigmoid* or *softmax*, U is a U-net like structure, composed of a series of downsampling (bottom-up) and upsampling (top-down) operations. In addition, skip connections between the corresponding bottom-up and top-down components are added to help learn A_s .

Dilated Residual Network

Dilated Residual Network (DRN) is composed of five Dilated Residual Modules (DRM) and two fully-connected (FC) modules. Each DRM has a residual unit, followed by a max-pooling layer and a dilated convolution layer.



Dilated Convolution

Dilated convolution operation $*_d$ is defined as,

$$(F *_d G)(n) = \sum_{m_1 + dm_2 = n} F(m_1)G(m_2), \forall m_1, m_2$$

where F is a feature map, G is a convolution kernel, and d is the dilation rate. With dilated convolution, the DNN's receptive field grows exponentially with layer depth such that it encodes more global knowledge.

Reference

- [1] Zhizheng Wu, et al. "Spoofing and countermeasures for speaker verification: A survey," *Speech Communication*, vol. 66, pp. 130–153, 2015.
- [2] Héctor delgado, et al. "Asvspoof 2017 version 2.0: meta-data analysis and baseline enhancements," in *Proc. Odyssey 2018 The Speaker and Language Recognition Workshop*, 2018, pp. 296–303.

Attention Heatmap Visualization

Effect of ϕ

Visualizations of A_s with different ϕ (from top to bottom): *Sigmoid*, *Tanh*, *SoftmaxF*, and *SoftmaxT*.

Sigmoid scales each dimension *independently* to range [0, 1], while *Softmax* scales each dimension *dependently* between to range [0, 1], implying only a few dimensions (either frequency bins or time frames) are activated and most are suppressed.

On the other hand, *Tanh* outputs in [-1, 1], and potentially losses useful information in S .

Effect of \bar{S}

Visualizations of A_s with and without \bar{S} : input feature map S (top), A_s learned without \bar{S} (middle) and with \bar{S} (bottom).

With \bar{S} , A_s strongly focuses on high frequency components of speech segments; without \bar{S} , A_s learns a summary of S .

Experimental Results

Systems	dev EER	eval EER	Diff.
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Version 2 dataset

AF(Sigmoid)+AF(SoftmaxT)	6.09	8.54	2.45
AF(Sigmoid)+AF(SoftmaxF)	6.37	8.80	2.43
AF(SoftmaxT)+AF(SoftmaxF)	6.39	8.98	2.59
AF(Sigmoid)-DRN(ReLU)	6.55	8.99	2.44
AF(SoftmaxT)-DRN(ReLU)	6.62	9.28	2.66
AF(SoftmaxF)-DRN(ReLU)	6.52	9.34	2.82
DRN(ELU)	7.49	10.16	2.67
AF(Tanh)-DRN(ReLU)	6.87	10.17	3.30
DRN(ReLU)	6.69	10.30	3.61
MDF(fusion)	-	6.32	-
qDFTspec	-	11.43	-
CQCC-GMM(CMVN)	9.06	12.24	3.18
i-vectors (Cosine Similarity)	8.99	14.77	5.78
i-vectors (Gaussian)	8.81	15.11	6.30
LCNN (our implementation)	6.47	16.08	9.61
Evolving RNN	18.7	18.20	-0.50
CQCC-GMM	12.08	29.35	17.27

Version 1 dataset

DLFS(fusion)	3.98	6.23	2.25
MDF(fusion)	-	6.54	-
LCNN	4.53	7.34	2.81
ConvRBM(fusion)	0.82	8.89	8.07
Multi-task	4.21	9.56	5.35
ResNet	10.95	16.26	5.31

*Our reported numbers are averaged over 8 runs.

Baselines

- CQCC-GMM
- i-vectors
- Light CNN

Single Systems

EERs of DRN and AF-DRN are reported. We can see that Attentive Filtering Network outperforms almost all previous work.

Fusions

Given visualizations of A_s we hypothesize that AF with different ϕ could be complementary, and as expected, fusing AF with *Sigmoid* and *Softmax* further reduces the EER.

Our fusion system provided a 30% relative improvement over the enhanced baseline system [2].