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#### Abstract

Voice assistant systems have become increasingly more commonplace, transcending personal use and expanding into corporate environments. Speaker recognition systems, specifically, often serve as another form of biometric authentication. Our research targets the vulnerabilities of such speaker recognition systems, particularly the X-Vector model. By adding small perturbations to audio input, we perform untargeted and targeted attacks, which aim to produce misclassification of audio and imitate another user, respectively.

## Background

#### • Voice Assistant Systems

- Often used to authenticate users via voice recognition
- Examples: Alexa, Google Home, Siri, Cortana, etc.

#### Adversarial Attacks

- Added perturbations to produce misclassifications
- Attacker intentionally designs these inputs to fool the model into making a mistake
- Shows that many modern machine learning algorithms can be broken





## **Adversarial Machine Learning Against Voice Assistant Systems**

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## **Materials and Methods**

- Implemented X-Vector model in Tensorflow • Extract Mel-frequency cepstral coefficients (MFCC) from the audio to serve as lower-dimensional representation
- Construct model using time-delay neural networks (TDNN) to capture contextual details of the audio
  - MFCC serve as input for TDNN
- Train probabilistic linear discriminant analysis (PLDA) classifier for robust speaker recognition
- **Equation:**  $X' = X + \in sign(\nabla x (-y*log(f(X))))$  Untargeted Attack • Misclassify audio sample as an incorrect speaker • Generate voice feature embeddings from the X-Vector model • Take the gradient of the cross-entropy cost function

- Add linear perturbation to original signal using fast gradient sign method (FGSM)
- Produce samples of varying strength, scaled by a given epsilon value Equation: (minimize)  $-y_t^*\log(f(X+\delta))+c||\delta||_2$ • Augment an audio sample to imitate a specific speaker • Iteratively modify perturbation according to the gradient descent
- Targeted Attack
- - direction where the possibility of target speaker increases • Minimize L2 norm of perturbation

# accuracy under attack.



We created a survey to determine the epsilon value at which the human ear could first detect our added perturbations. The results are displayed in the pie chart above.

**Targeted Attack:** The bar graph depicts the average noise level in decibels (dB) for adversarial samples targeting each speaker.

## **Conclusions and Future Work**

## • Untargeted attack:

- Targeted attack:
- Future Work:
  - impulse response

[1] D. Snyder, D. Garcia-Romero, G. Sell, D. Povey and S. Khudanpur, "X-Vectors: Robust DNN Embeddings for Speaker Recognition," 2018 IEEE International Conference on Acoustics, Speech and Signal Processing (ICASSP), Calgary, AB, 2018, pp. 5329-5333 [2] Li, Zhuohang & Shi, Cong & Xie, Yi & Liu, Jian & Yuan, Bo & Chen, Yingying. (2020). Practical Adversarial Attacks Against Speaker Recognition Systems. 9-14. 10.1145/3376897.3377856.





#### Discussion

**Untargeted Attack:** The left plot shows different epsilon values and their corresponding noise level in decibels (dB). The right plot shows different epsilon values and the corresponding PLDA

 $\circ \epsilon = 1E-5$  distortion was inaudible to human ear but

decreased PLDA accuracy by ~20%

Earliest discernable distortion occurred at  $\varepsilon = 0.001$ , with the PLDA accuracy decreased by almost 70%

• Average noise level of -21 dB,  $\sim \epsilon = 0.0316$  in untargeted

Launch over-the-air attacks and account for the room

• Better disguise attacks to produce greater unpredictability

References