





Vulnerabilities of Voice Assistants at the Edge: From Defeating Hidden Voice Attacks to Audio-based Adversarial Attacks

Yingying (Jennifer) Chen

Professor, Electrical and Computer Engineering Department Associate Director, WINLAB Director, Data Analysis and Information Security (DAISY) Lab Rutgers University, New Brunswick, NJ, USA yingche@scarletmail.rutgers.edu http://www.winlab.rutgers.edu/~yychen/

IEEE ICNP Workshop AIMCOM2

October 13, 2020





Wireless Information Network Laboratory (WINLAB)







- □ Industry-university research center founded in1989
 - Focus on wireless technology
- □ Hosting world-class researchers
 - ✤ 20 faculties from different departments
 - ✤ 45 PhD students
- Active research directions:
 - Mobile ad hoc networks (MANET) for tactical applications
 - Mesh network protocols
 - Delay tolerant networks (DTN)
 - Software defined networks
 - Mobile content delivery
 - Wireless network security







Open-Access Research Testbed for Next-Generation Wireless Networks (ORBIT)



ORBIT nodes







Control room

400 - USRP open access research testbed
Funded by NSF since 2003 with \$12M

Research Applications:

- ✤ 5G mm wave
- Mobile edge cloud and future mobile Internet
- Healthcare IT and Internet of Things (IoT)
- Mobile sensing and user behavior recognition
- Network coding and spectrum management
- Vehicular networking





Cloud Enhanced Open Software Defined Mobile Wireless Testbed for City-Scale Deployment (COSMOS)

- □ Funded by NSF PAWR for \$22M in 2018 for deploying 5G network testbed
- Led by Rutgers and collaborating with Columbia University, New York University and University of Arizona
- □ Focus on 5G technologies
 - Ultra-high bandwidth and low latency wireless communication
- □ Tightly coupled with edge cloud computing



- Deployment in New York City
- ✤ 9 Large sites and 40 Medium sites
- ✤ 200 small nodes to support edge computing

Research Applications:

- Ultra-high bandwidth, low latency, and powerful edge computing
- Future mobile Internet and mobile edge cloud
- Healthcare IT and Internet of Things (IoT)
- AR and VR





- Fiber connection to Rutgers, GENI/I2, NYU
- Interaction with smart community





DAISY Data Analysis and Information SecuritY Lab

Defeating Hidden Audio Channel Attacks on Edge Voice Assistants - via Audio-Induced Surface Vibrations





Motivation

Widely deployed voice controllable systems (VCS) at the edge

- Convenient way of interaction
- Integrated into many platforms

Mobile phones (e.g., Siri and Google Now)





Smart appliances





stand-alone assistants



WINLAB

Fundamental vulnerabilities due to the propagation properties of sound
Emerging hidden voice commands
Recognizable to VCS
Incomprehensible to humans



Hidden Voice Command

Attacks the disparities of voice recognition between human and machine

Iteratively shaping their audio features to meet the requirements:

✤Understandable to VCSs

Hard to be perceived by the users

Attack model

- Internal attack embedded in media and played by the target device
- External attack played via a loudspeaker in the proximity

browse evil.com

ΓGERS



Related Work

Defend acoustic attacks based on audio information

Voice authentication models

Only relying on speech audio features is vulnerable to hidden voice commands

Speech vocal features (e.g.,)

□Speaker liveness detection



Restricted application scenarios by either requiring the microphone to be held close to mouth or additional dedicated hardware

A multi-modality authentication framework is highly desirable

to provide enhanced security:

Audio sending modality + vibration sensing modality





Basic Idea

Many VCS devices (e.g., smartphones and voice Basic Idea: utilizing the vibration signatures of the voice command to detect hidden voice commands

Unique audio-induced surface vibrations captured by the motion sensor are hard to forge

Two modes for capturing noticeable speech impact on motion sensors based on playback



Front-end playback

Replay Device in Cloud Service



Motion Sensor S

Back-end playback





10

Capturing Voice Using Motion Sensors

Shared surface between loudspeaker and microphone
Low sampling rate motion sensors (e.g., < 200Hz)
Nonlinear vibration responses
Distinct vibration domain

$$f_{alias} = |f - Nf_s|, N \in Z_s$$



FRS



Why Vibration?

Existing speech/voice recognition methods based on audio domain voice vocal features

- Hidden voice commands designed to duplicate these audio domain features by iteratively modify a voice command
- □Audio-induced surface vibrations
 - An additional sensing domain, distinct to audio

The vibration domain approach can work in conjunction with the audio domain approach to more effectively detect the hidden voice commands.

physical vibrations, motion sensors)





System Overview







Vibration Feature Derivation

□Unique and hard to forge

- Statistical features in time and frequency domains
- Deriving Acoustic Features from Motion Sensor Data
 - ≻MFCC
 - ≻Chrome vectors

Nonlinear relationship between audio features and vibration features





Vibration Feature Derivation

□Unique and hard to forge vibration features

- Statistical features in time and frequency domains
- Deriving Acoustic Features from Motion Sensor Data
 - **≻**MFCC



Feature Selection Based on Statistical Analysis







Hidden Voice Command Detection

Supervised Learning-based method

- ✤ Simple Logistic
- ✤Support Vector Machine
- Random Forest
- Random Tree
- Unsupervised learning-based method
 - k-means/k-medoids based methods
 - Calculating the Euclidean distance of the voice command samples to the cluster centroid
 - Not require much training







Experimental Setup

- Front-end playback setup
 - ✤4 different smartphones
 - ✤On table
 - Held by hand
 - Placed on sofa
- Backend playback setup
 - Imitated cloud service device
 - Prototype on Raspberry Pi
- 10 voice commands, 5 speakers
- □ 13,000 vibration data traces

FGERS

- ♦6500 benign commands
- ♦6500 hidden voice commands





17

Performance Evaluation Unsupervised-learning



Up to 99% accuracy for both frontend and backend setups to differentiate normal commands from hidden voice commands





Performance Evaluation

□Partial playback to reduce delay

Front-end playback setup

	Note 4	G3	Nexus 6	S6
Replay all	100%	99.10%	100%	85.70%
Replay 1s	100%	89.10%	99.90%	85.60%
Replay 0.5s	99.90%	85.20%	95.90%	85%

Back-end playback setup

	Note 4	G3	Nexus 6	S6
Replay all	99.90%	97.90%	93.40%	76%
Replay 1s	92.9	99.10%	92.40%	75.90%
Replay 0.5s	88.5	90.20%	90.50%	73.80%

Various mobile device usage scenarios of frontend playback setup

	Tabla	Held	Placed	80%vol.	2x speed	
	Table	in hand	on sofa	on table	on table	
Kmed	100%	87.30%	100%	100%	88.30%	
Kmea	100%	87.30%	100%	100%	85.20%	





- Demonstrate that hidden voice commands can be detected by their speech features in the vibration domain
- Derive the unique vibration features (statistical features in the time and frequency domains and speech features to distinguish hidden voice commands from normal commands
- Develop both supervised and unsupervised learning-based systems to detect hidden voice commands
- Implemented the proposed system in two modes: frontend playback and backend playback







DAISY Data Analysis and Information SecuritY Lab







What's Speaker Recognition?

□Speaker Recognition (SR)



Attack Chances on Speaker Recognition

□Trend in Speaker Recognition

Adopting Deep Neural Networks (DNNs) for better performance ^[1]

DNNs are vulnerable to *adversarial examples* ^[2, 3]

Recognized as Stop



Benign IBBH Ign Input

Recognized as Speed Limit 4 Gibbon



Perturbationarsariade erangial Example

Mitchell McLaren, Yun Lei, and Luciana Ferrer. 2015. Advances in deep neural network approaches to speaker recognition. In IEEE ICASSP 2015.
Ian J Goodfellow, Jonathon Shlens, and Christian Szegedy. 2014. Explaining and harnessing adversarial examples. arXiv:1412.6572 (2014).
Eykholt, Kevin, et al. "Robust physical-world attacks on deep learning visual classification." Proceedings of the IEEE Conference on Computer Vision and Pattern Recognition. 2018.

23





Limitation of Existing Attacks

□ Speaker Recognition Pipeline







Limitation of Existing Attacks

Conventional Attacks

- Replay attack, synthesis attack, voice conversion attack
- Pros: injected via physical channel
- **Cons**: can be defended by modern SR models ^[4, 5]



[4] Hong Yu, Zheng-Hua Tan, Yiming Zhang, Zhanyu Ma, and Jun Guo. 2017. DNN filter bank cepstral coefficients for spoofing detection. IEEE Access 5 (2017), 4779–4787.
[5] Zhizheng Wu, Tomi Kinnunen, Eng Siong Chng, Haizhou Li, and Eliathamby Ambikairajah. 2012. A study on spoofing attack in state-of-the-art speaker verification: the telephone speech case. In IEEE APSIPA ASC 2012. 1–5.





Limitation of Existing Attacks

Adversarial Attack

- Leverage adversarial examples
- Pros: strong, can fool state-of-the-art model
- Cons: success in digital domain, sensitive to overthe-air distortions



against state-of-the-art speaker recognition system



First practical adversarial attack against multi-class SR system

Use the estimated room impulse response to launch over the air attack

Implement gradient-based algorithms to make the attack unnoticeable

Evaluate on a public dataset of 109 English speakers







... ...





28

















Target Model

□ X-vector ^[6]

The state-of-the-art DNN-based multi-class speaker



[6] David Snyder, Daniel Garcia-Romero, Gregory Sell, Daniel Povey, and Sanjeev Khudanpur. 2018. X-vectors: Robust dnn embeddings for speaker recognition. In IEEE ICASSP 2018.





Problem Formulation



Attack Overview







Room Impulse Response Estimation

\BoxRoom Impulse Response (RIR) – h(t)

✤ Model the transfer function between the played audio x(t) and the received audio y(t)

$$y(t) = x(t) \otimes h(t)$$

RIR estimation

• Play an excitation signal $x_e(t)$

$$x_{e}(t) = \sin\left(\frac{2\pi f_{1}T}{\ln(\frac{f_{2}}{f_{1}})} \left(e^{\frac{t}{T}\ln(\frac{f_{2}}{f_{1}})} - 1\right)\right)$$

- Record the response $y_e(t)$
- ♦ Estimate RIR, where f(t) is the time-reversal of $x_e(t)$

$$h(t) = y_e(t) \otimes f(t)$$





Room Impulse Response Estimation

□ Preliminary Experiment

♦ f = 20 - 20kHz, T = 5s

- Measured Mean Square Error (MSE)
 - ➢Recorded & Predicted = 0.112

➢Original & Recorded = 0.84







Adversarial Example Generation

Untargeted Attack

Due to the local linearity of DNN models, a linear perturbation is sufficient for untargeted attacks ^[7]:

$$\begin{cases} X' = X + \delta \\ \delta = \epsilon sign\left(\nabla_X J\left(X, y\right)\right) \\ J\left(X, y\right) = -y \cdot log(P) \end{cases}$$

Digital untargeted adversarial example

$$X' = X + \epsilon sign\Big(\nabla_X \big(-y \cdot log(f(X)) \big) \Big)$$

✤ Practical untargeted adversarial example $X' = X + \epsilon sign \Big(\nabla_X \big(-y \cdot log(f(X \otimes h)) \big) \Big)$

[7] Ian J Goodfellow, Jonathon Shlens, and Christian Szegedy. 2014. Explaining and harnessing adversarial examples. arXiv:1412.6572 (2014).





Adversarial Example Generation

Targeted Attack

Adversarial example targeting at label y_t can be generated through solving an **optimization problem**:

minimize $||\delta||_2$, s.t. $f(X + \delta) = y_t$

Lagrangian relaxation:

minimize
$$-y_t \cdot log(f(X + \delta)) + c||\delta||_2$$

- ✤ Apply gradient descent to find the optimal δ^*
- Digital targeted adversarial example

$$X' = X + \delta^*$$

Practical targeted adversarial example

minimize $-y_t \cdot log(f((X + \delta) \otimes h)) + c||\delta||_2$





Experimental Methodology

- CSTR VCTK Corpus
- Total 44217 utterances spoken by 109 English speakers with various accents, training & testing ratio = 4:1

Baseline Model

- ✤ 30 dimensional MFCC with frame length of 25 ms
- Pretrained X-vector model provided in Kaldi^[8]

Evaluation Metrics

- Speaker Recognition Accuracy (%)
- ✤ Attack Success Rate (%)
- Distortion Metric (dB)



Evaluation of Digital Attacks

Digital Untargeted Attack

Test set : 8896 audio files

Attack Strength (i.e., ϵ)	No Attack	10 ⁻⁵	10^{-4}	10 ⁻³	10 ⁻²	10 ⁻¹
Speaker Recognition Accuracy (%)	92.81	84.71	41.33	12.11	2.23	1.37
Attack Success Rate (%)	_	8.73	55.47	86.95	97.60	98.52
Average Distortion (dB)	_	-89.06	-69.15	-49.24	-29.33	-9.41
				1		

Digital Targeted Attack

Tested on all original-target speaker combinations (total 109*108 pairs)

• /				
Attack Strength (i.e., c)	0.4	0.2	0.1	0.05
Attack Success Rate (%)	77.64	86.05	93.27	96.01
Average Distortion (dB)	-34.22	-32.43	-29.66	-25.94





Evaluation of Practical Attack

DExperimental Setup

- Two realistic scenarios: office & apartment
- 10 digital/practical targeted adversarial example tested in each scenario





(a) Office

(b) Apartment

	Playing digital	Playing practical		
	adversarial examples	adversarial examples		
Office	0%	50%		
Apartment	10%	50%		





Audio Samples

□Making Speaker #1 recognized as Speaker #20

- Original audio
 - Recognized as Speaker #1
- Practical adversarial audio
 - Misrecognized as Speaker #20
 - ➤ Measured distortion: -42.35dB
- ✤ Genuine speech from Speaker #20











Take-aways

- We demonstrate a practical and systematic adversarial attack against DNN-based speaker recognition systems
- Apply gradient-based algorithms to launch both untargeted and targeted attacks
- Integrate the estimated RIR into the adversarial example generation for a more practical attack
- Conduct extensive experiment in both digital and realworld settings





Future work: Security Issues on Voice Recognition Systems at the edge - Attacker could control your smart home



<u>Future work: Security Issues on Augmented Reality (AR) System</u> - Attacker could control your 'reality'





Thanks to my collaborators and students