



Robust Speaker Recognition in the Cross-Microphone Condition

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Doug Reynolds

NIST Speaker Recognition Workshop

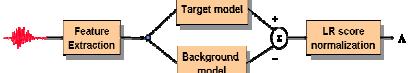
27 June 2006

This work was sponsored by the Department of Defense under Air Force contract F19628-00-C-0002. Opinions, interpretations, conclusions, and recommendations are those of the authors and are not necessarily endorsed by the United States Government.



Outline

- System Overview
 - Core systems and development data
- Cross-channel 2006
 - Feature Mapping
 - SVM-GSV+NAP
 - Multi-Feature SVM-GLDS+NAP
- Performance Analysis
 - Telephone vs. Xchan 2005 and 2006
 - Per-microphone results
- Conclusions



```

graph LR
    Input(( )) --> FE[Feature Extraction]
    FE --> Target[Target model]
    FE --> BG[Background model]
    Target --> LR[LR score normalization]
    BG --> LR
    LR --> A[A]
    style Input fill:none,stroke:none
    style FE fill:#e0e0e0
    style Target fill:#e0e0e0
    style BG fill:#e0e0e0
    style LR fill:#e0e0e0
  
```

System Overview

- Second year on NIST cross-microphone task
- Focused on three spectral based detectors
- Main emphasis on channel compensations

| Core Detectors | System | Features | Classifier | Znorm | Tnorm | Chan. Comp. |
|----------------|----------|-----------------------|------------|-------|-------|-------------|
| | GMM-LFA | MFCC | GMM | 200 | 300 | LFA |
| | SVM-GSV | GMM mean SuperVectors | SVM | | | NAP |
| | SVM-GLDS | MFCC+LPCC | SVM | | | NAP |

| Development Data | System | Background | Znorm | Tnorm | Chan. Comp |
|------------------|----------|--|-------|------------|--------------------------|
| | GMM-LFA | SWB2, SRE04 | SWB2 | SRE04, FSH | SWB2, SRE05-XC |
| | SVM-GSV | ubm=SWB2 svm=FSH | | | SWB2, SRE05-XC |
| | SVM-GLDS | FSH-ENG | | | SWB2, SRE05-XC |
| | FUSION | Cross-Validation on system scores from SRE05 | | | |

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Cross-channel 2006

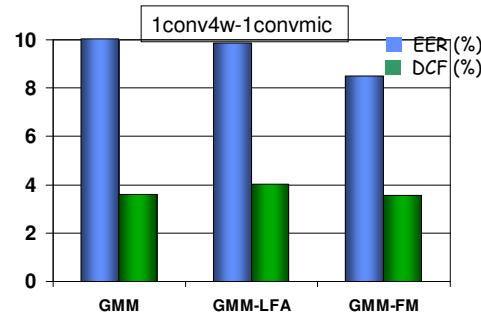
- Leverage the channel/session compensation developed in SRE-06 telephone systems:
 - Latent Factor Analysis (GMM-LFA)
 - Nuisance Attribute Projection (SVM-GSV, SVM-GLDS)
 - Feature-Mapping with convmic models (PostEval)
- Factor Loading Matrix (LFA) / Projection Matrix (NAP)
 - Trained with pooled telephone and cross-channel data
 - Limited cross-channel development data**
 - 97 Speakers in SRE-2005 X-Channel corpus
 - 47 Speakers contained both X-Channel and Telephone data
- Based on development data, telephone trained LFA matrix used for GMM-LFA
 - Too little data per speaker for good gender-dependent estimation

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Feature Mapping Post Evaluation System

- Added microphone dependent models to feature mapper
 - Trained with SRE05 xchan data
 - Channels c1-c8
 - Gender dependent
- Total of 22 models
 - 6 telephone models (cell, cordless, regular)
- Appears to be better than LFA for 1c/1c
 - Perhaps using limited xchan data more effectively
- Currently coupling with other systems

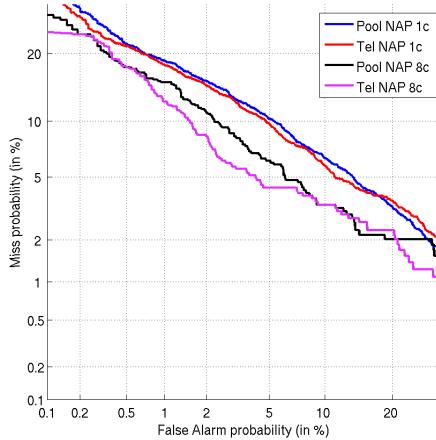


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SVM-GSV+NAP

- Eval strategy:**
 - Pool NAP: Pool telephone data with xchan microphone data
 - Design NAP projection to eliminate all variation
 - Cons: Development data reused for cross-validation (fusion, thresholds)
- Alternate strategy:**
 - Tel NAP: Use models with default telephone session NAP projection
 - Cons: No modeling of xchan microphones



Conclusion: Not much difference for SVM-GSV between pooled and telephone NAP projection at minDCF.

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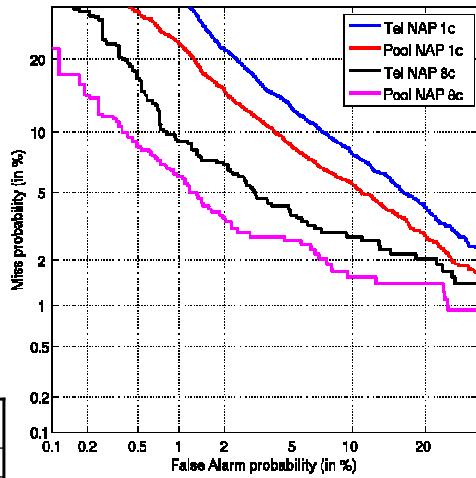
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Multi-Feature SVM-GLDS+NAP

- Strategies for NAP:
 - Pool telephone data with xchan microphone data
 - Telephone only
- Analysis:
 - Pooled NAP works significantly better
 - NAP interacts differently with different feature sets
 - LPCCs are not as good as MFCCs under mismatch; need NAP to make them fuse well

| | MFCC EER (%) | LPCC EER (%) | Fuse EER (%) |
|--------------|-----------------|-----------------|-----------------|
| 1c, Tel NAP | 10.04 | 14.07 | 8.84 |
| 1c, Pool NAP | 9.22 | 10.34 | 6.88 |
| 8c, Tel NAP | 4.04 | 7.91 | 4.19 |
| 8c, Pool NAP | 3.72 | 5.27 | 2.90 |



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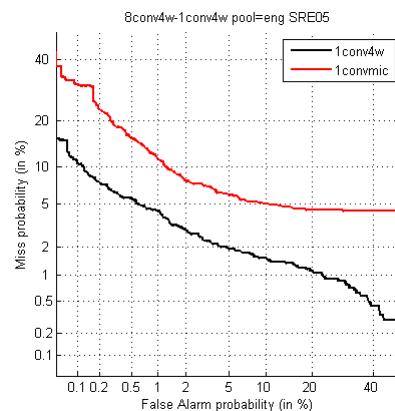
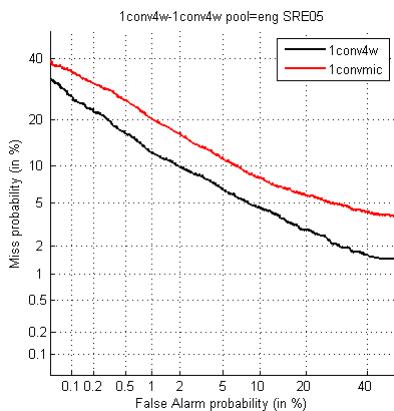
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Performance Analysis Telephone vs. Xchan 2005

- In 2005 answer key bug made it look like microphone data was harder than it really was
- Gap in performance still there, but more reasonable
 - Systems applied in 2005 were not tuned for microphone data



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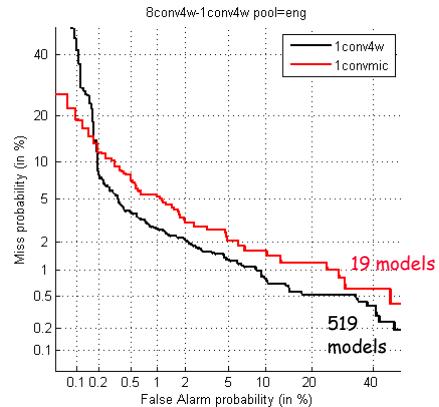
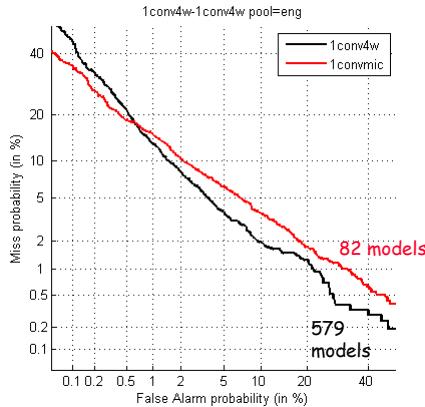
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1conv4w DETS are from 1c/1c trials not just xchan telephone trials. Using 2005 systems.



Performance Analysis Telephone vs. Xchan 2006

- Limited tests to ENG since xchan data is almost all ENG
- Same systems used for 1conv4w and 1convmic tests
 - Focus was on effect of changing input not different core system combinations
- Relatively small loss in accuracy between telephone and microphone inputs in the aggregate



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1conv4w DETS are from 1c/1c trials not just xchan telephone trials

**Performance Analysis
Xchan Microphones**

0. 2. 1. 4. 11 NIST SRE 26-27 June 2006

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**Performance Analysis
Per-Microphone (1conv4w)**

10
8
6
4
2
0

1conv4w
82 spkrs

EER (%)
DCF (%)

Microphone: c0, c1, c2, c3, c4, c5, c6, c7, c8

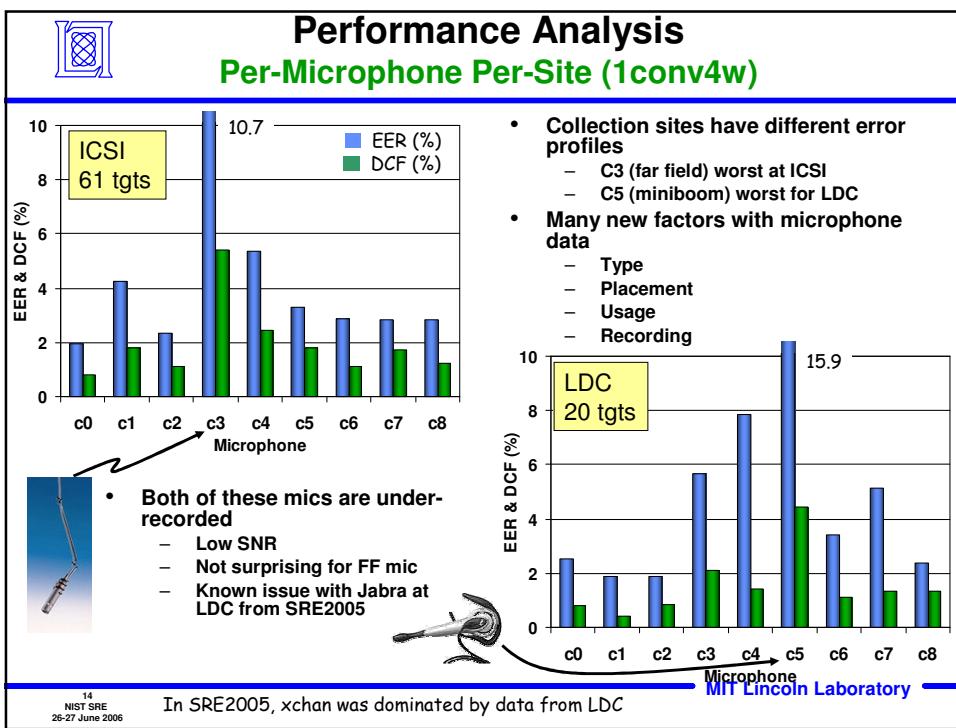
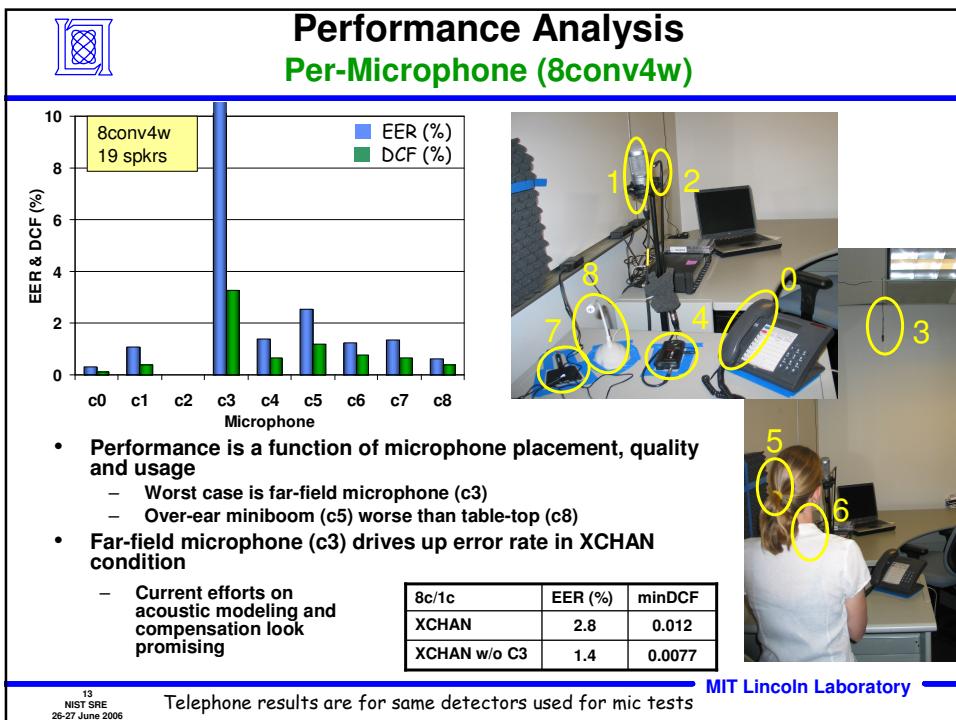
- Performance is a function of microphone placement, quality and usage
 - Worst case is far-field microphone (c3)
 - Over-ear miniboom (c5) worse than table-top (c8)
- Far-field microphone (c3) appears to drive up error rate in XCHAN condition
- There is also variability with XCHAN collection site (LDC and ICSI)

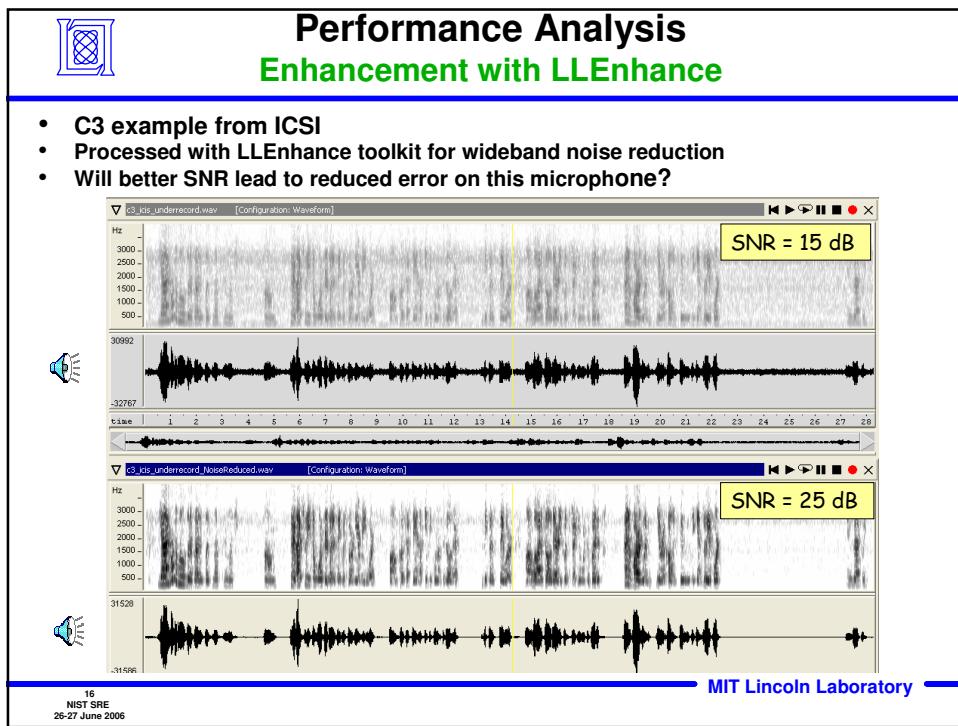
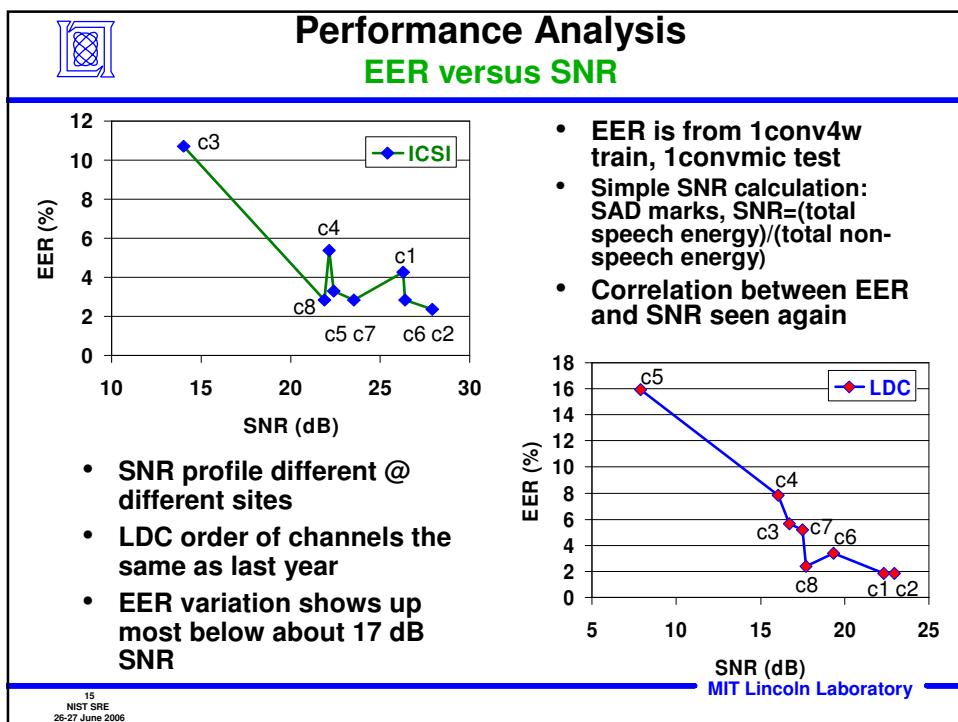
| 1c/1c | EER (%) | minDCF |
|--------------|---------|--------|
| XCHAN | 5.8 | 0.023 |
| XCHAN w/o C3 | 3.9 | 0.018 |

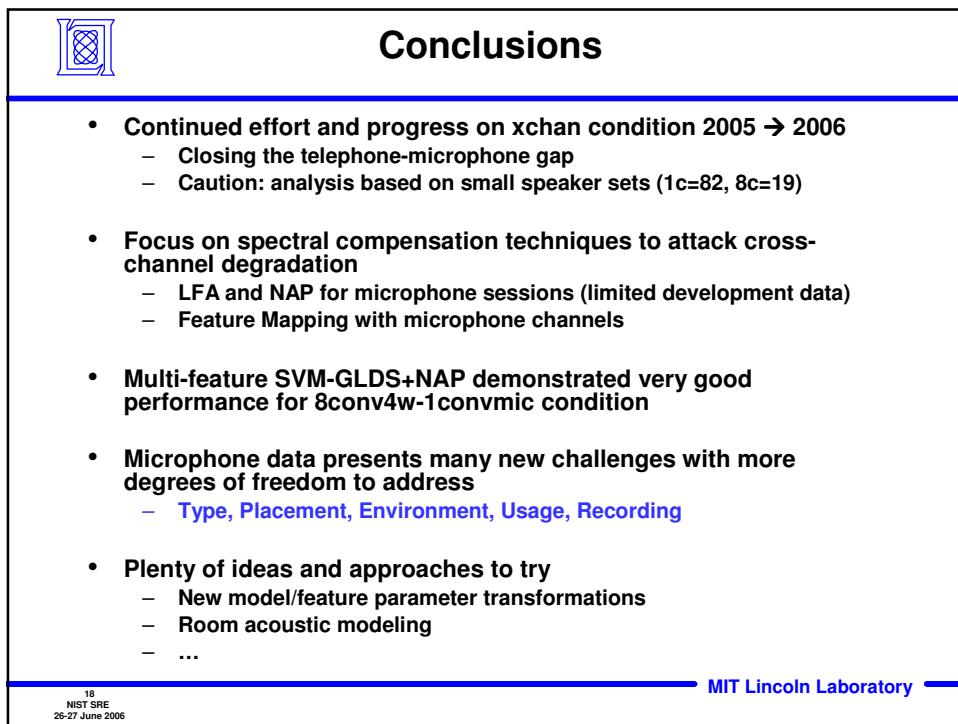
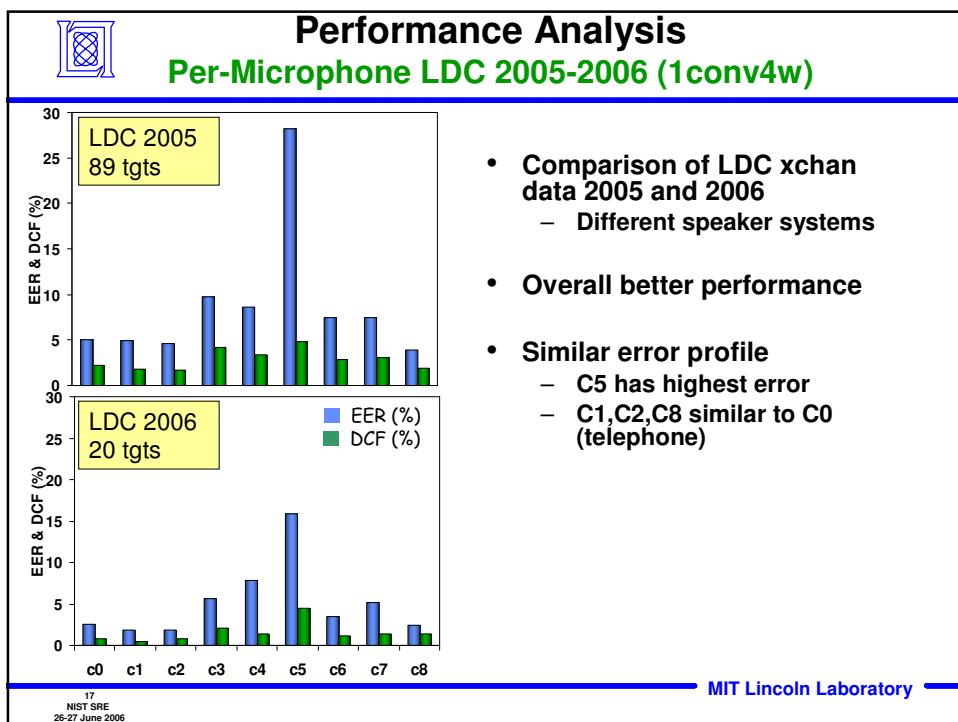
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Telephone results are for same detectors used for mic tests

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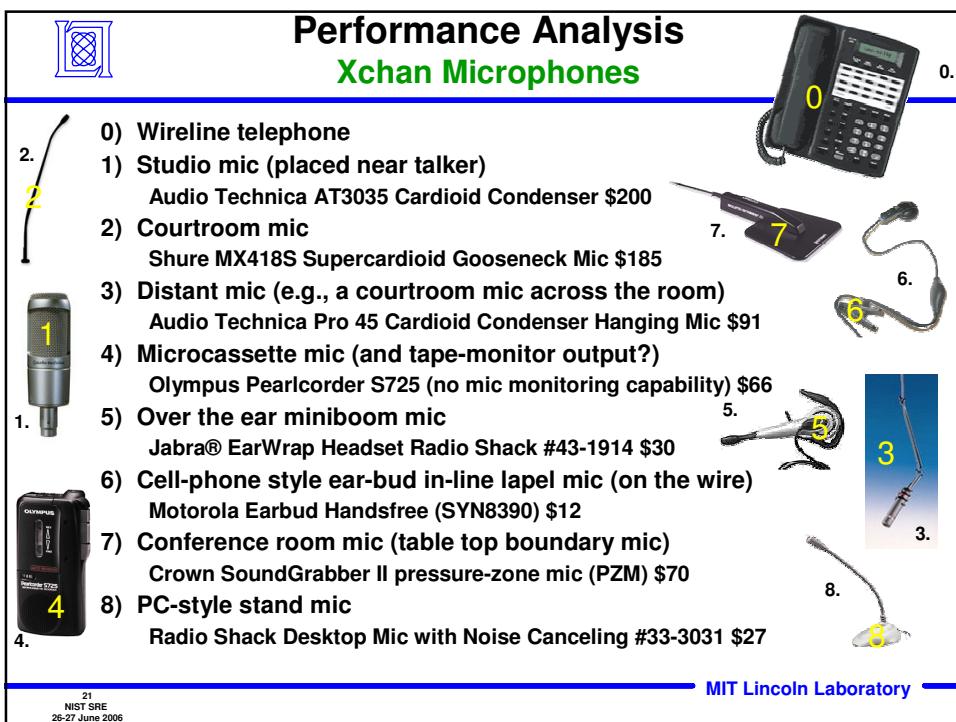


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Xchan Discussion

- Microphone data presents many new challenges with more degrees of freedom to address
 - **Type**: transducer characteristics
 - **Placement**: where the microphone is placed relative to speaker and room characteristics (coupling)
 - **Acoustic environment**: room characteristics (size, surfaces, noise sources, etc.)
 - **Usage**: how the speaker (mis)uses the microphone
 - **Recording**: how the transducer signal is recorded
- Telephone
 - Feedback (listener or sidetone)
 - Active communication channel vs passive recording
 - Handset induces better placement of microphone
- Need to converge on key dimensions
 - Current setup focuses on type (some on placement)
 - Are placement and acoustic environment more important factors?



Performance Analysis
Xchan Microphones

0. 0.

2. 2.

1) Studio mic (placed near talker)
Audio Technica AT3035 Cardioid Condenser \$200

2) Courtroom mic
Shure MX418S Supercardioid Gooseneck Mic \$185

3) Distant mic (e.g., a courtroom mic across the room)
Audio Technica Pro 45 Cardioid Condenser Hanging Mic \$91

4) Microcassette mic (and tape-monitor output?)
Olympus Pearlcorder S725 (no mic monitoring capability) \$66

5) Over the ear miniboom mic
Jabra® EarWrap Headset Radio Shack #43-1914 \$30

6) Cell-phone style ear-bud in-line lapel mic (on the wire)
Motorola Earbud Handsfree (SYN8390) \$12

7) Conference room mic (table top boundary mic)
Crown SoundGrabber II pressure-zone mic (PZM) \$70

8) PC-style stand mic
Radio Shack Desktop Mic with Noise Canceling #33-3031 \$27

1. 1.

4. 4.

7. 7.

6. 6.

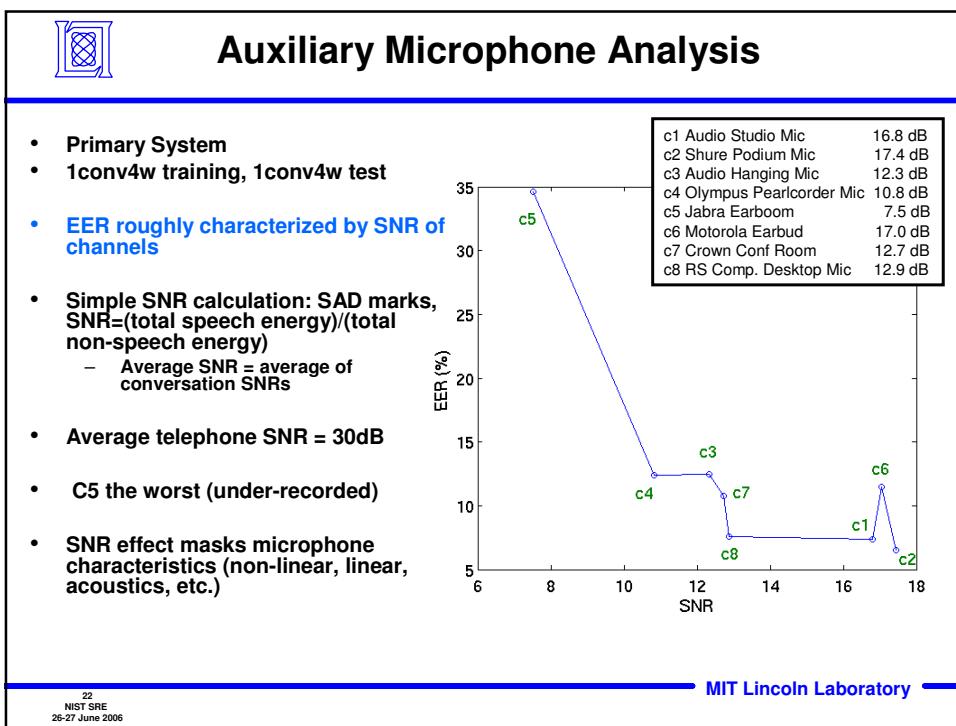
5. 5.

3. 3.

8. 8.

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Auxiliary Microphone Analysis

• Primary System

• 1conv4w training, 1conv4w test

• **EER roughly characterized by SNR of channels**

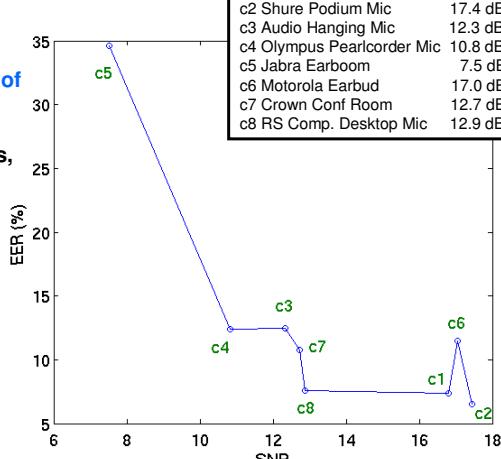
• Simple SNR calculation: SAD marks, $SNR = (\text{total speech energy}) / (\text{total non-speech energy})$
– Average SNR = average of conversation SNRs

• Average telephone SNR = 30dB

• C5 the worst (under-recorded)

• SNR effect masks microphone characteristics (non-linear, linear, acoustics, etc.)

c1 Audio Studio Mic 16.8 dB
c2 Shure Podium Mic 17.4 dB
c3 Audio Hanging Mic 12.3 dB
c4 Olympus Pearlcorder Mic 10.8 dB
c5 Jabra Earboom 7.5 dB
c6 Motorola Earbud 17.0 dB
c7 Crown Conf Room 12.7 dB
c8 RS Comp. Desktop Mic 12.9 dB



The graph plots EER (%) on the y-axis (ranging from 5 to 35) against SNR on the x-axis (ranging from 6 to 18). The data points are connected by a line, showing a general downward trend. The points are labeled with codes: c5 (at ~7.5 SNR, ~33 EER), c4 (at ~11 SNR, ~12 EER), c3 (at ~12.5 SNR, ~13 EER), c8 (at ~13 SNR, ~6 EER), c7 (at ~13.5 SNR, ~11 EER), c1 (at ~16.5 SNR, ~7 EER), and c6 (at ~17 SNR, ~13 EER). A legend on the right lists the microphone types and their corresponding SNR values.

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Feature Domain Compensation Feature Mapping

