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Testing the Newest Phonexia Speaker Recognition System With Forensic Eval 01 and Further Corpora

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Introduction

The performance of automatic speaker comparison software has increased dramatically with the introduction of artificial neural networks. To include the automatic approach in a forensic phonetic analysis, forensic practitioners need to be aware of the system's performance and potential pitfalls. As a continuation of previous testing efforts performed and described by Enzinger & Morrison (2016, 2019) and Jessen (2019), Phonexia's newest neural network models were tested and results are described hereafter. A real-case dataset, designed specifically for system testing, was used (Morrison & Enzinger 2016) and widely accepted performance metrics such as EER and various types of Cllr's were obtained (described in (Morrison & Enzinger 2016).



Figure 1: Tippett plot (with precision) xl4 (baseline)



Figure 3: DET plots for all models (model xl4 (mean normalised) shows the best results

Further Work & Outlook

• Further testing should be conducted, e.g. investigating the performance of the systems on low-resource language such as Swiss-German and synthetic voices

Methods

The system test followed the testing guidelines set by Morrison & Enzinger (2016). The dataset contains files with two different



recording conditions that vary in their quality and content (details provided in Morrison & Enzinger (2016)). Additional training data was further provided alongside the test dataset, which was used in the current test to further enhance the system. MFCC's were calculated as features and embedded with deep neural networks into so-called x-vectors (Snyder et al. 2018) for comparison. The tested models are similar to the SID-Beta4 version as described in detail in Jessen et al. (2019). Two enhancement methods were used, namely Mean Normalization and FAR (False Acceptance Rate)–Calibration. The calculation of the Cllr's and EER's was done in Matlab using a script provided by Geoffrey Morrison.

Figure 2: Tippett plot (with precision) I4 (baseline)

- Preliminary tests with Swiss-German were very promising
- Testing of systems should occur on a regular basis, since the models and hence the performance of the systems change rapidly
- Further testing should be done with speech messages of e.g. WhatsApp and Snapchat

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Results						
	Cllr _{pooled}	CIIr _{mean}	95% CI	CIIr _{min}	CIIr _{cal}	EER
XI4 Baseline	0.219	0.184	0.535	0.085	0.134	0.021
XI4 10 % FAR	0.184	0.149	0.559	0.0784	0.106	0.019
XI4 10% FAR + Mean Normalisation	0.127	0.107	0.686	0.050	0.077	0.015
XI4 Mean Normalisation	0.089	0.055	0.0659	0.045	0.043	0.012
L4 Baseline	0.194	0.179	0.902	0.111	0.083	0.028

Conclusion

In comparison to the older versions of Phonexia (tested in Jessen 2019), the system performance has increased. Comparing the obtained test results to previous evaluations of other systems (Morrison & Enzinger, 2019) shows that Phonexia's newest XI4 model is scoring the best, respectively is showing the lowest error rates and log likelihood ratios. However, further testing should be conducted, e.g. investigating the performance on low-resource languages such as Swiss-German.