

# **2022 International Workshop on Acoustic Signal Enhancement (IWAENC 2022)**

**Bamberg, Germany  
5 – 8 September 2022**



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*mh acoustics*

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<sup>1</sup>*Paderborn University, Germany*  
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<sup>3</sup>*National Institute of Advanced Industrial Science and Technology, Japan*  
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*HEAD acoustics, Germany*

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<sup>1</sup>Hörzentrum Oldenburg, Germany

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