

Audio Engineering Society

121st Convention Papers 2006

October 5-8, 2006
San Francisco, California, USA

Volume 1 of 3

Printed from e-media with permission by:

Curran Associates, Inc.
57 Morehouse Lane
Red Hook, NY 12571
www.proceedings.com

ISBN: 978-1-60423-663-7

Some format issues inherent in the e-media version may also appear in this print version.

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2006

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