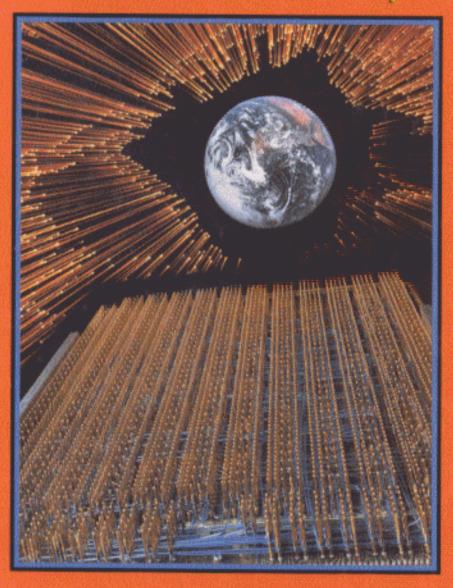
# DIGITAL SIGNAL PROCESSING

An Overview of Basic Frinciples



Jack Cartinhour

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