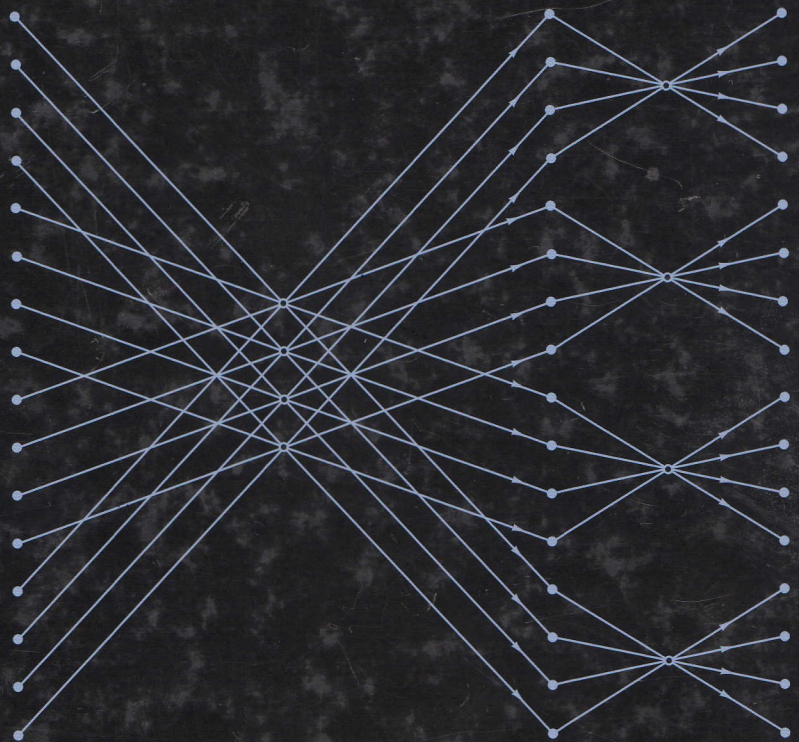


DIGITAL SIGNAL PROCESSING

Principles, Algorithms,
and Applications

Fourth Edition



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