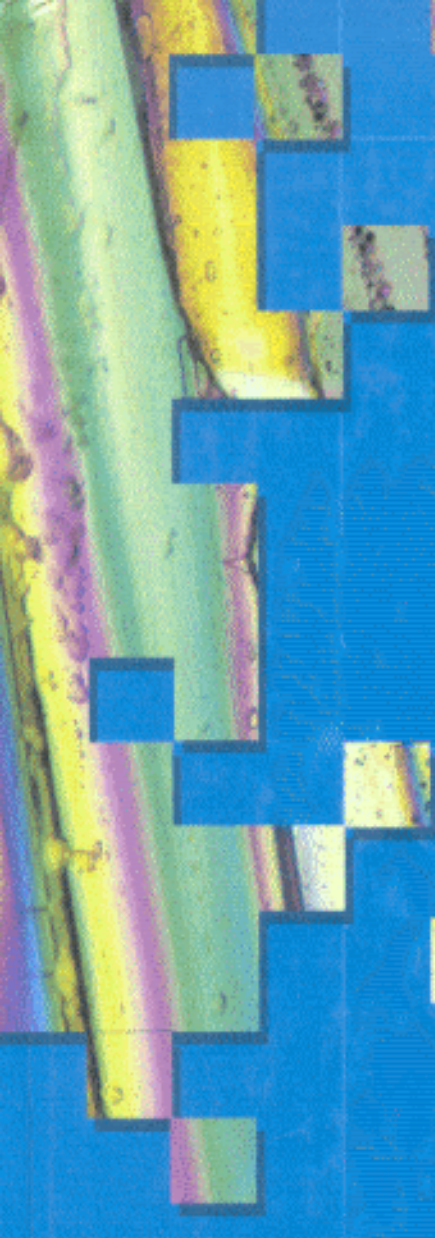




Fundamentals of Digital Signal Processing

Joyce Van de Vegte

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