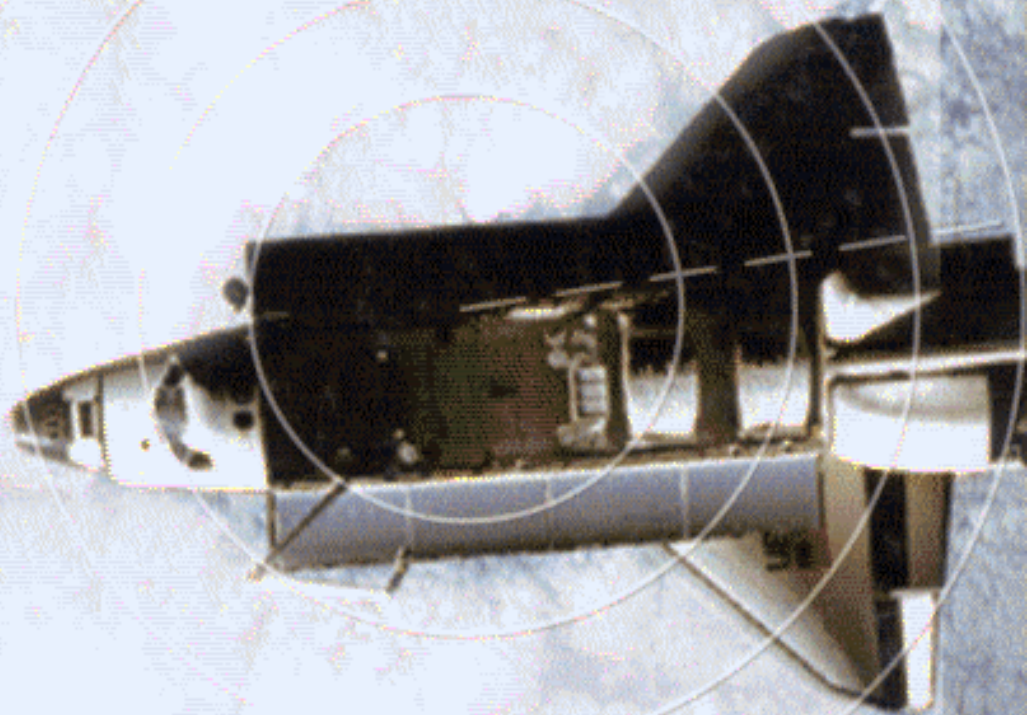


Digital Signal Processing

with Examples in MATLAB®



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CRC PRESS

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