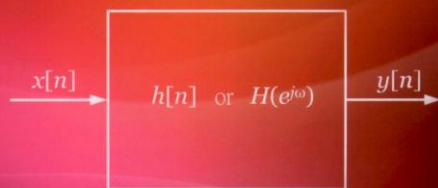


Digital Signal Processing: Foundations, Transforms and Filters, with Hands-on MATLAB Illustrations



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