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ประวัติผู้เขียน

นายวรพจน์ พัฒนวิจิตร เกิดเมื่อวันที่ 6 พฤศจิกายน พ.ศ. 2515 ที่กรุงเทพมหานคร สำเร็จการศึกษาระดับปริญญาตรี หลักสูตรวิศวกรรมศาสตรบัณฑิต สาขาวิศวกรรมไฟฟ้า จากภาควิชาวิศวกรรมไฟฟ้า คณะวิศวกรรมศาสตร์ จุฬาลงกรณ์มหาวิทยาลัย ในปีการศึกษา 2537 และเข้าศึกษาต่อในหลักสูตรวิศวกรรมศาสตรมหาบัณฑิต สาขาวิศวกรรมไฟฟ้า ภาควิชาวิศวกรรมไฟฟ้า คณะวิศวกรรมศาสตร์ จุฬาลงกรณ์มหาวิทยาลัย เมื่อปี พ.ศ. 2537